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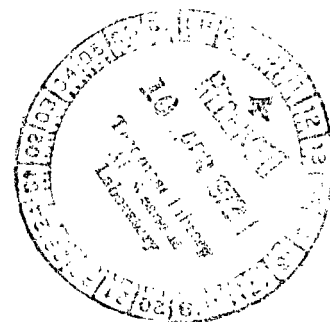
DESIGN OF RECURSIVE DIGITAL FILTERS
HAVING SPECIFIED PHASE
AND MAGNITUDE CHARACTERISTICS

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DESIGN OF RECURSIVE DIGITAL FILTERS HAVING SPECIFIED PHASE AND MAGNITUDE CHARACTERISTICS

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SUMMARY

A method for a computer-aided design of a class of optimum filters, having specifications in the frequency domain of both magnitude and phase, is described. The method, an extension to the work of Steiglitz, uses the Fletcher-Powell algorithm to minimize a weighted squared magnitude and phase criterion. Results using the algorithm for the design of filters having specified phase as well as specified magnitude and phase compromise are presented.

INTRODUCTION

Recursive filters, wherein the output sequence is both a function of the input as well as past output samples, are commonly used in digital signal processing and analysis. Such digital filters in many applications offer distinct advantages of precision and versatility over their continuous or analog counterparts. There exist a number of design procedures for implementing digital filters (see ref. 1) each one of which strives to attain some analogy between discrete and continuous systems. Transform methods such as the matched- z , bilinear- z , and standard- z which lead to specific property invariances are available (see ref. 2) to the designer familiar with continuous filter design.

For frequency-domain synthesis (see refs. 3 and 4), realization is normally by means of cascade or parallel combinations of pole and zero pairs in the complex plane. The synthesis problem is, in fact, reduced to one of approximation since the filter topology is generally specified. In none of the available design procedures, which can yield filters having excellent magnitude-frequency characteristics, however, do the resultant filters, in themselves, have particularly useful phase characteristics. Indeed, in striving for particular magnitude characteristics by using any of the available design methods, there is no control over the filter phase properties.

In practice, it is often desirable to specify a digital filter in the frequency domain by its phase (see ref. 5) or even a compromise between magnitude and phase. The procedure in this paper meets these requirements through the use of an iterative computer-aided design leading to an optimum set of parameters for a specified filter topology and is an extension of the technique described by Steiglitz (see ref. 6) for determining the optimum coefficients of a cascade filter having magnitude specifications alone. The extension makes possible the design of a new class of digital filters having the prescribed phase characteristics.

SYMBOLS

A	filter multiplier
D_k^i	denominator of i th stage of $H(z)$ at Ω_k
E_k^M	magnitude error at Ω_k
E_k^ϕ	phase error at Ω_k
\vec{e}_k	error vector at Ω_k
$\partial \vec{e}_k / \partial A$	derivative of error vector at Ω_k with respect to zero frequency gain
f_k	frequency at k th specification point, Hz
f_s	sampling frequency, Hz
$H(z)$	unity gain discrete transfer function
$ H_k $	magnitude of $H(z)$ at Ω_k
\overline{H}_k	conjugate of $H(z)$ at Ω_k
$\partial H_k / \partial \vec{p}$	gradient vector of magnitude of $H(z)$ at Ω_k with respect to parameter vector
$I()$	imaginary part of quantity
i, \dots, N	denotes filter stage

\vec{J}_k	Jacobian at Ω_k , $\left[A^* \frac{\partial H_k }{\partial \vec{p}} \quad \frac{\partial \phi_k}{\partial \vec{p}} \right]$
k	sample point
M_k	specification magnitude at Ω_k
N_k^i	numerator of i th stage of $H(z)$ at Ω_k
\vec{p}	parameter vector
\vec{p}_i	set of filter parameters for the i th stage, a_i , b_i , c_i , and d_i
$q_1^i(k)$	first system state of i th stage at k th sample point
$q_2^i(k)$	second system state of i th stage at k th sample point
$R(\)$	real part of quantity
$u_i(k)$	input to i th stage at k th sample point
V	criterion functional, that is, $V(A, \vec{p})$
V_k	criterion functional at Ω_k , that is, $V_k(A, \vec{p})$
\hat{V}	reduced criterion functional, that is, $V(A^*, \vec{p})$
$\partial V / \partial A$	slope of criterion functional with respect to zero frequency gain
$\partial V_k / \partial \vec{e}_k$	gradient vector of criterion functional at Ω_k with respect to error vector at Ω_k
\vec{W}_k	weighting matrix at Ω_k
W_k^M	magnitude weighting at Ω_k
W_k^ϕ	phase weighting at Ω_k
$w^i(k)$	dummy variable of i th stage at k th sample point

$Y(z)$	digital filter discrete transfer function
$y_i(k)$	output of i th stage at k th sample point
z	transform variable
z_k	discrete transform variable at Ω_k , $e^{j\pi\Omega_k}$
θ_k	specification phase at Ω_k , radians
λ	collective phase weight
ϕ_k	phase of $H(z)$ at Ω_k , radians
$\partial\phi_k/\partial\vec{p}$	gradient vector of phase of $H(z)$ at Ω_k with respect to parameter vector
Ω_k	fractional frequency at k th specification point

An asterisk on a symbol denotes an optimum value. A circumflex denotes optimization with respect to A . A superscript T denotes the transpose.

DISCUSSION

The Filter Form

The fundamental advantages of the N -stage cascade canonical form of recursive digital filter whose signal flow graph is shown in figure 1 and which is described by the product operator

$$Y(z) = A \prod_{i=1}^N \frac{1 + a_i z^{-1} + b_i z^{-2}}{1 + c_i z^{-1} + d_i z^{-2}} \quad (1)$$

$$Y(z) = AH(z)$$

are (1) its relative insensitivity to perturbations in the denominator coefficients, an important consideration in digital filters, especially of high order and particularly where finite register lengths (see ref. 1) are involved; (2) its simplicity of implementation; and (3) the simplicity of factoring the filter operator to determine its roots. This form has found extensive application in practical filters for signal processing, and a version employing serial arithmetic (ref. 7) is commercially available.

For completeness, an alternative description of the filter is given in terms of the system states q_1^i and q_2^i and clearly demonstrates the recursive nature of the filter. The set of difference equations describing the filter and required in developing a computer algorithm is presented. Thus, for the i th stage in figure 1 at the k th sample point

$$w^i(k) = A_i u_i(k) - c_1 q_1^i(k) - d_1 q_2^i(k)$$

$$q_1^i(k+1) = w^i(k)$$

$$q_2^i(k+1) = q_1^i(k)$$

$$y_i(k) = w^i(k) + a_1 q_1^i(k) + b_1 q_2^i(k)$$

where

$$u_i(k) = y_{i-1}(k)$$

is the input to the i th stage and is identical to the output of the $(i-1)$ stage and

$$A_i = \begin{cases} A & (i = 1) \\ 1 & (i \neq 1) \end{cases}$$

The Synthesis Problem

The design problem considered in this paper can be stated as follows: When the magnitude and phase specifications (M_k and θ_k , respectively) at the k th fractional Nyquist frequencies $\Omega_k = 2f_k/f_s$ (where f_s is the sampling frequency in Hz) are known, determine the set of optimum parameters \vec{p}^* of an N -stage cascade filter having the form of equation (1) so that the resultant digital filter will have a minimum sum squared magnitude and phase error for all specified frequencies.

By constraining the filter topology, the optimum synthesis problem becomes one of parametric optimization with respect to a given criterion of fit. The composite criterion which can weight the magnitude and phase requirements independently and as functions of frequency is chosen as the inner product

$$V(A, \vec{p}) = \sum_k \langle \vec{e}_k, \vec{W}_k \vec{e}_k \rangle = \sum_k V_k \quad (2)$$

where

$$\vec{e}_k = \begin{bmatrix} A |H_k| - M_k \\ \phi_k - \theta_k \end{bmatrix} = \begin{bmatrix} E_k^M \\ E_k^\phi \end{bmatrix}$$

is the error vector and

$$\vec{W}_k = \begin{bmatrix} W_k^M & 0 \\ 0 & \lambda W_k^\phi \end{bmatrix}$$

is the diagonal weighting matrix. Clearly, $V(A, \vec{p})$ is a nonlinear function of the parameter vector $\vec{p} = (a_1, b_1, c_1, d_1, \dots, a_N, b_N, c_N, d_N)^T$, which involves the $4N$ filter coefficients, and of the filter multiplier A .

The Minimization Algorithm

Through formal differentiation of the criterion function (eq. (2)) with respect to the multiplier A , the minimization procedure can be slightly simplified to that of finding the minimum of a reduced functional $\hat{V}(\vec{p}) = V(A^*, \vec{p})$ involving only $4N$ parameters. Thus

$$\frac{\partial V}{\partial A} = \sum_k \left\langle \frac{\partial \vec{e}_k}{\partial A}, \frac{\partial V_k}{\partial \vec{e}_k} \right\rangle = 2 \sum_k \begin{bmatrix} |H_k| W_k^M & 0 \end{bmatrix} \vec{e}_k$$

and $\partial V / \partial A = 0$ implies

$$2 \sum_k |H_k| W_k^M (A^* |H_k| - M_k) = 0$$

or

$$A^* = \frac{\sum_k |H_k| W_k^M M_k}{\sum_k |H_k|^2 W_k^M} \quad (3)$$

An additional necessary condition for existence of an extremum is that the gradient vector be zero; thereby, the optimum parameter vector \vec{p}^* is obtained. From equation (2)

$$\frac{\partial \hat{V}}{\partial \vec{p}} = 2 \sum_k \left\langle \vec{J}_k, \vec{W}_k \vec{e}_k \right\rangle \quad (4)$$

where the $(4N \times 2)$ Jacobian \vec{J}_k is

$$\vec{J}_k^T = \nabla_{\vec{p}} \vec{e}_k = \begin{bmatrix} A^* \frac{\partial |H_k|}{\partial \vec{p}} & \vdots & \frac{\partial \phi_k}{\partial \vec{p}} \end{bmatrix}^T \quad (5)$$

Clearly, each element of the gradient vector is the sum of two weighted functions of the magnitude and phase error. By writing

$$|H_k|^2 = H_k \bar{H}_k$$

where \bar{H}_k is the conjugate of H_k evaluated at the fractional frequency Ω_k , it is readily shown (see ref. 6), where \vec{p}_i is the set of filter parameters for the i th stage, that

$$\frac{\partial |H_k|}{\partial \vec{p}_i} = \frac{1}{|H_k|} R \left(\bar{H}_k \frac{\partial H_k}{\partial \vec{p}_i} \right)$$

For the cascaded filter topology in terms of the elements of \vec{p}_i ,

$$\frac{\partial |H_k|}{\partial a_i} = |H_k| R \left(\frac{z_k^{-1}}{N_k^i} \right)$$

$$\frac{\partial |H_k|}{\partial b_i} = |H_k| R \left(\frac{z_k^{-2}}{N_k^i} \right)$$

$$\frac{\partial |H_k|}{\partial c_i} = -|H_k| R \left(\frac{z_k^{-1}}{D_k^i} \right)$$

and

$$\frac{\partial |H_k|}{\partial d_i} = -|H_k| R \left(\frac{z_k^{-2}}{D_k^i} \right)$$

where, with $z_k = e^{j\pi\Omega_k}$,

$$N_k^i = N^i(z_k) = 1 + a_i z_k^{-1} + b_i z_k^{-2}$$

and

$$D_k^i = D^i(z_k) = 1 + c_i z_k^{-1} + d_i z_k^{-2}$$

By letting

$$H_k = |H_k| e^{j\phi_k}$$

it follows that

$$\phi_k = I(\log_e H_k)$$

whence

$$\frac{\partial \phi_k}{\partial \vec{p}} = I\left(\frac{\partial}{\partial \vec{p}} \log_e H_k\right) = I\left(\frac{1}{H_k} \frac{\partial H_k}{\partial \vec{p}}\right)$$

which takes on a particularly simple form for the cascade topology. For the i th stage parameters, in fact,

$$\frac{\partial \phi_k}{\partial a_i} = I\left(\frac{z_k^{-1}}{N_k^i}\right)$$

$$\frac{\partial \phi_k}{\partial b_i} = I\left(\frac{z_k^{-2}}{N_k^i}\right)$$

$$\frac{\partial \phi_k}{\partial c_i} = -I\left(\frac{z_k^{-1}}{D_k^i}\right)$$

and

$$\frac{\partial \phi_k}{\partial d_i} = -I \left(\frac{z_k^{-2}}{D_k^i} \right)$$

The special case of a one-stage ($N = 1$) filter is illustrated. Here

$$H_k = A \frac{1 + az_k^{-1} + bz_k^{-2}}{1 + cz_k^{-1} + dz_k^{-2}}$$

$$\hat{V} = \sum_k \left(A^* |H_k| - M_k \right)^2 W_k^M + \lambda \sum_k (\phi_k - \theta_k)^2 W_k^\phi$$

and

$$\frac{\partial \hat{V}}{\partial a} = 2 \sum_k \left(E_k^M W_k^M \frac{\partial |H_k|}{\partial a} + \lambda E_k^\phi W_k^\phi \frac{\partial \phi_k}{\partial a} \right) = \sum_k \left[Q_k^M R \left(\frac{z_k^{-1}}{N_k^i} \right) + \lambda R_k^\phi I \left(\frac{z_k^{-1}}{N_k^i} \right) \right]$$

Similarly,

$$\frac{\partial \hat{V}}{\partial b} = \sum_k \left[Q_k^M R \left(\frac{z_k^{-2}}{N_k^i} \right) + \lambda R_k^\phi I \left(\frac{z_k^{-2}}{N_k^i} \right) \right]$$

$$\frac{\partial \hat{V}}{\partial c} = \sum_k \left[Q_k^M R \left(\frac{z_k^{-1}}{D_k^i} \right) + \lambda R_k^\phi I \left(\frac{z_k^{-1}}{D_k^i} \right) \right]$$

$$\frac{\partial \hat{V}}{\partial d} = \sum_k \left[Q_k^M R \left(\frac{z_k^{-2}}{D_k^i} \right) + \lambda R_k^\phi I \left(\frac{z_k^{-2}}{D_k^i} \right) \right]$$

where

$$Q_k^M = 2E_k^M W_k^M |H_k|$$

and

$$R_k^\phi = 2E_k^\phi W_k^\phi$$

are the weighted errors. It is obvious that the frequency intervals of the input data (specifications) need not be uniform and may, in fact, be intentionally unequal to allow for nonuniform frequency weighting.

Complementary Root Reflection and Stability

In deriving the frequency response of a discrete operator by letting z_k lie on the unit circle Γ , it is possible to take advantage of a unique property of the discrete transform pertaining to its magnitude when a root lying outside the unit circle is imaged or reflected into the unit circle. It is easy to show that the magnitude of a phasor $z - z_0$, where z_0 is a root of the discrete transform lying outside the unit circle, is equal to

$$|z - z_0| = |z_0| \left| z - \frac{1}{z_0} \right|; z \in \Gamma$$

Since z_0 has been assumed to be outside the unit circle, $1/z_0$ must be inside, the term $|z_0|$ correcting for magnitude changes. Thus, if in the optimization procedure a pole should stray outside the unit circle and thereby lead to an unstable filter, root reflection guarantees stability with no magnitude change. There is no analogous simple identity for the phase of a reflected root. Experience with the procedure has shown that provided the design requirements can be met by means of a stable filter, that is, that a feasible solution exists, an optimum will indeed be found through repeated application of root reflection.

The Computer Algorithm

A complete listing of the filter design algorithm, which is an adaptation of the program written by Steiglitz, is given in the appendix. The main program is termed STGZ3 which calls four principal subroutines: (1) FUNCT performs the functional and gradient computation for each iteration as well as putting out the final optimum parameters and plots, (2) FLPWL is a Fletcher-Powell conjugate gradient routine, (3) INSIDE computes root reflection, and (4) ROOTS determines the poles and zeros of the filter. Single-precision arithmetic has been employed.

When minimization of the functional has been attained in the first pass or the minimization algorithm has iterated 300 times, a test is made to ascertain that all the roots are within the unit circle, a necessary requirement for the poles for stability reasons and for the zeros to insure minimum phase. If the design should result in an unstable

configuration, the roots are reflected about the unit circle and minimization is resumed in a second pass. If a minimum does indeed exist and all the roots then lie within the unit circle, the program computes and prints out the frequency response and commences plotting.

Minimization is deemed to be achieved when the absolute difference in functionals between successive iterations $\epsilon = |\hat{V}_{\text{new}} - \hat{V}_{\text{old}}|$ or the norm of the gradient vector falls below preassigned limits. Convergence is generally fast for magnitude or phase filters but can be very slow for the case of compromise filters.

When the design specifications cannot be met after LIM iterations (see appendix), the program will stop; this situation indicates that the optimum could not be found and the resultant characteristic which may be unusable is plotted. Generally, feasible designs have been determined in less than 2000 iterations.

Minimization of the criterion function does not guarantee determination of a global minimum but rather determination of a local minimum. Depending upon the parameter vector utilized for initialization of the algorithm computation, different minima may be achieved. Experience has shown that stage-by-stage optimization, that is, utilization of the i th-stage optimum parameter vector as the initial parameter vector for the $(i + 1)$ stage of an N -stage filter, yields lower minimum values of the criterion function than does single-pass optimization.

APPLICATIONS

Linear-Phase Filter

This example considers a digital filter having application as a phase discriminator with a linear phase characteristic and arbitrary magnitude characteristic and is shown in figure 2. In this example all magnitude weights were set to zero and all phase weights to unity, the multiplier A being arbitrarily made unity since it has no effect on the phase characteristic.

The phase requirements were $\theta_k = 1 - 2\Omega_k$ ($0 \leq \Omega_k \leq 1$), and a two-stage filter was specified. When an initial parameter vector $\vec{p} = (0, 0, 0, 0.25, 0, 0, 0, 0)^T$ was used, the algorithm converged to the optimum, with $\epsilon = 10^{-4}$, in 52 iterations and a Control Data 6600 computer time of 14 seconds. The optimum parameter values computed were to four places

$$A = 1.0$$

$$a_1 = 0 \quad b_1 = -0.9871 \quad c_1 = 0 \quad d_1 = 0.0395$$

$$a_2 = 0 \quad b_2 = -0.9871 \quad c_2 = 0 \quad d_2 = -0.0127$$

It is interesting to note that the phase requirements were met to within 0.008π radian for approximately 95 percent of the frequency range.

Constant-Phase Filters

Two cases were considered to obtain filters having constant phases of $-\pi/2$ and $\pi/2$ radians over a frequency range $0.3 \leq \Omega_k \leq 0.7$. As in the previous case, the form of the magnitude characteristic was of no concern; hence, zero magnitude weighting was specified. With the same initial parameter state used in the previous example, the first case (lag network) optimized in 1673 iterations and 42 seconds to yield a hyperbolic magnitude characteristic and phase errors of less than 0.0003π radian throughout the specified band.

The computed parameters for the lag case were

$$A = 1.0$$

$$a_1 = 0.5580 \quad b_1 = -0.1857 \quad c_1 = -0.4752 \quad d_1 = 0.0363$$

$$a_2 = 0.5580 \quad b_2 = -0.1857 \quad c_2 = -0.3712 \quad d_2 = -0.5686$$

The positive phase filter (lead network), however, took only 165 iterations and 17 seconds to yield the desired phase characteristic with errors nowhere exceeding 0.001π radian in the specified band.

The optimum filter parameters for this second case were determined to be

$$A = 1.0$$

$$a_1 = -0.4768 \quad b_1 = -0.1548 \quad c_1 = 0.5022 \quad d_1 = -0.1082$$

$$a_2 = -0.4768 \quad b_2 = -0.1548 \quad c_2 = 0.4515 \quad d_2 = -0.2008$$

It is noted that for both cases, the phase weights outside the specified band were set to zero, and thereby allowed for arbitrary phase in these regions. Figures 3(a) and 3(b) show the resultant frequency characteristics for the lag and lead cases, respectively, of two-stage filters. The combination of the two filters, although they have antagonistic magnitude characteristics, suggests the possibility of a phase-splitting digital network.

Limited-Band Constant-Gain Linear-Phase Filter

The third example demonstrates a compromise design of a digital filter having constant-magnitude and linear-phase characteristics, over a limited frequency band, typical of phase discriminators. Here, except for $\lambda = 0$, the specifications were stated as

$$M_k = \begin{cases} 1 & (0.3 \leq \Omega_k \leq 0.7) \\ 0 & (\text{Elsewhere}) \end{cases}$$

$$\theta_k = \begin{cases} 1 - 2\Omega_k & (0.3 \leq \Omega_k \leq 0.7) \\ 0 & (\text{Elsewhere}) \end{cases}$$

Equal error and frequency weights were employed and the effects of changes in λ are shown in figure 4 for a two-stage design. Figure 4(a) shows the case of $\lambda = 0$, that is, a magnitude-only filter being specified, and coincidentally yields the linear-phase-filter characteristic derived in the first example. (See fig. 2.) Figures 4(b) and 4(c) show the magnitude and phase characteristics for the cases of $\lambda = 10$ and $\lambda = 1000$, respectively. The increasing weight on phase and resultant degradation in the magnitude characteristic are shown. The optimum parameters were

$\lambda = 0$:

$$A = 0.2063$$

$$a_1 = 0.0000 \quad b_1 = -1.0000 \quad c_1 = 0.0000 \quad d_1 = 0.1539$$

$$a_2 = 0.0000 \quad b_2 = -1.0000 \quad c_2 = 0.0000 \quad d_2 = 0.1539$$

$\lambda = 10$:

$$A = 0.3658$$

$$a_1 = -0.9754 \quad b_1 = 0.7300 \quad c_1 = 0.4529 \quad d_1 = 0.7211$$

$$a_2 = 0.8632 \quad b_2 = 0.5632 \quad c_2 = -0.6119 \quad d_2 = 0.7443$$

$\lambda = 1000$:

$$A = 0.4232$$

$$a_1 = -1.1739 \quad b_1 = 0.8489 \quad c_1 = 0.7596 \quad d_1 = 0.6691$$

$$a_2 = 1.1739 \quad b_2 = 0.8489 \quad c_2 = -0.7596 \quad d_2 = 0.6691$$

Low-Pass Zero-Phase Filter

The fourth example considers a compromise filter, having two and three stages, with specifications that are intentionally conflicting. A filter described by

$$M_k = \begin{cases} 1.0 & (0.0 \leq \Omega_k < 0.5) \\ 0.5 & (\Omega_k = 0.5) \\ 0.0 & (0.5 < \Omega_k \leq 1.0) \end{cases}$$

$$\theta_k = \begin{cases} 0 & (0.0 \leq \Omega_k \leq 0.5) \\ \text{Unspecified} & (\text{Elsewhere}) \end{cases}$$

is specified.

Figures 5 and 6 show the results for the two- and three-stage designs, respectively, with figures 5(a) and 6(a) showing the magnitude-only ($\lambda = 0$) case. The degradation in the magnitude characteristics when greater emphasis is placed on the phase specifications is evident in figures 5(b) and 6(b) for $\lambda = 10$ and in figures 5(c) and 6(c) for $\lambda = 1000$. Comparison of figure 6 with figure 5 demonstrates the improvement brought about by increasing the number of stages. The optimum parameters for the two-stage filter were

$\lambda = 0$:

$$A = 0.1196$$

$$a_1 = 1.0240 \quad b_1 = 1.0000 \quad c_1 = -0.1713 \quad d_1 = 0.7676$$

$$a_2 = 1.0240 \quad b_2 = 1.0000 \quad c_2 = -0.5324 \quad d_2 = 0.2286$$

$\lambda = 10$:

$$A = 0.4879$$

$$a_1 = 0.2018 \quad b_1 = 0.6684 \quad c_1 = 0.3560 \quad d_1 = 0.4612$$

$$a_2 = 0.6597 \quad b_2 = 0.4335 \quad c_2 = 0.0806 \quad d_2 = 0.7671$$

$\lambda = 1000$:

$$A = 0.5343$$

$$a_1 = 0.0205 \quad b_1 = 0.7169 \quad c_1 = -0.0836 \quad d_1 = 0.6255$$

$$a_2 = 0.6286 \quad b_2 = 0.7905 \quad c_2 = 0.2123 \quad d_2 = 0.6681$$

The optimum parameters for the three-stage filter were

$\lambda = 0$:

$$A = 0.0510$$

$$a_1 = 0.8537 \quad b_1 = 1.0000 \quad c_1 = -0.1068 \quad d_1 = 1.0000$$

$$a_2 = 0.8537 \quad b_2 = 1.0000 \quad c_2 = -0.4046 \quad d_2 = 0.5990$$

$$a_3 = 0.8537 \quad b_3 = 1.0000 \quad c_3 = -0.6799 \quad d_3 = 0.2069$$

$\lambda = 10$:

$$A = 0.5109$$

$$a_1 = 1.3302 \quad b_1 = 0.5515 \quad c_1 = -0.1731 \quad d_1 = 0.8097$$

$$a_2 = 0.6844 \quad b_2 = 0.7157 \quad c_2 = 1.1880 \quad d_2 = 0.5850$$

$$a_3 = -0.0373 \quad b_3 = 0.7012 \quad c_3 = 0.3825 \quad d_3 = 0.5262$$

$\lambda = 1000$:

$$A = 0.4515$$

$$a_1 = 1.5107 \quad b_1 = 0.5286 \quad c_1 = -0.1771 \quad d_1 = 0.8972$$

$$a_2 = 0.5825 \quad b_2 = 0.7490 \quad c_2 = 1.3094 \quad d_2 = 0.4191$$

$$a_3 = -0.1663 \quad b_3 = 0.7485 \quad c_3 = 0.2002 \quad d_3 = 0.6393$$

A three-stage design of this example is used to demonstrate the existence of two distinct local minima, dependent upon the initial parameter vector. In the first case, a single-pass optimization was accomplished with $\vec{p} = (0, 0, 0, 0.25, 0, 0, 0, 0)^T$ for the initial parameter vector and resulted in the optimum filter shown in figure 6(a). In the second case, a stage-by-stage optimization was accomplished by utilizing the optimum parameter vector from a two-stage design for the initial parameter vector of a three-stage design and resulted in the optimum filter shown in figure 7. Comparison of these results demonstrates the existence of two distinct local minima, the stage-by-stage minimization yielding superior results.

CONCLUDING REMARKS

A method has been developed for a computer-aided design of cascade canonical digital filters having prescribed magnitude or phase characteristics or a compromise between the two. The method, which uses an unconstrained minimization algorithm, allows for arbitrary error and frequency weighting. Representative designs of phase and compromise filters have demonstrated the utility of the technique. Although convergence is generally fast for magnitude phase filters, it may be slow for the case of compromise filters.

Langley Research Center,
National Aeronautics and Space Administration,
Hampton, Va., February 17, 1972.

APPENDIX

PROGRAM LISTING

This appendix contains a program listing written for the Control Data 6600 computer at the Langley Research Center, Hampton, Virginia, and is an adaptation of that written by Kenneth Steiglitz at Princeton University for the design of specified magnitude-only filters.

```

PROGRAM STG73(INPUT,OUTPUT,TAPE5=INPUT,TAPE6=OUTPUT,PUNCH)      0013
EXTERNAL FUNCT                                                    0014
DIMENSION H(184),X(16),G(16)                                     0015
COMMON/RAW/W(100),Y(100),M,PHASED(100),ALAMDA,FR,WTM(100),      0016
C WTP(100),KTOP                                                    0017
COMMON/RAW1/ICALL,KCALL,LIM
CALL CALCOMP                                                    0019
CALL LFR(Y)                                                      0020
WRITE(6,51)                                                       0021
51 FORMAT(* INPUT DATA*)                                         0022
M=0                                                                0023
30 M=M+1                                                           0024
READ(5,21)W(M),Y(M),PHASED(M),WTM(M),WTP(M)                     0025
21 FORMAT(5F10.5)                                                 0026
WRITE(6,22)M,W(M),Y(M),PHASED(M),WTM(M),WTP(M)                 0027
22 FORMAT(* I=*,I3,* W=* F 7.4,* Y=* F 7.4,* PHASED=* F 7.4,    0028
C * WTM=* F 7.4,* WTP=* F 7.4)                                     0029
IF(W(M).LT.1.00)GOTO30                                           0030
DO 15 J=1,16                                                       0031
15 X(J)=0.00                                                       0032
X(4)=.25                                                           0033
99 READ(5,60)L,LIM,FST,FPS,HMAX,ALAMDA,FR,KTOP                  0035
60 FORMAT(2I5,5F10.5,I10)                                         0036
IF(FR.LT..001)FR=1.                                               0037
N=4*L                                                              0038
IF(FST,5) 999,FRR                                                 0039
488 CONTINUE
WRITE(6,61)L,LIM,FST,FPS,HMAX,FR,ALAMDA
61 FORMAT(* L=*,I3,* LIM=*,I5,* FST=* F10.5,* FPS=* F10.5,      0042
C * HMAX=* F10.5,* FREGRANGE=* F10.5,* LAMBDA=* F10.5)          0043
ICALL=0
99 KCALL=0
CALL FLPL(FUNCT,N,X,F,G,FST,FPS,FC,IER,H)
CALL ROOTS(N,X)                                                    0045
CALL INSIDE(N,X,KFLAG)                                             0046
WRITE(6,26)IER,KFLAG,ICALL,KCALL
26 FORMAT(* IER=*,I5,* KFLAG=* I5,* ICALL=* I5,* KCALL=* I5)
IF(KCALL.GT.300)GO TO 98
IF((KFLAG.NE.0.00,IER.NE.0).AND.(ICALL.LE.LIM))GO TO 98
CALL ROOTS(N,X)                                                    0050
ICALL=-10                                                           0051
CALL FUNCT(N,X,F,G,HMAX)                                           0052
GOTO 99                                                             0053
999 CALL CALPLT(CO,CO,999)                                         0054
STOP                                                                0055
END                                                                0056

```

APPENDIX – Continued

	SUBROUTINE FUNCT(N,X,F,G,HMAX)	0057
000010	DIMENSION CMFSA(200),PHASEFX(200),AMAG(200),PHE(100),PHA(100)	0058
000010	DIMENSION XPL(200),YPL(200),CX(200),CY(200)	0059
000010	DIMENSION H(164),X(16),C(16),YHT(100),F(100)	0060
000010	COMPLEX ZQPAR,ZC,ZZCUR,ZZCUP2	0061
000010	COMPLEX Z(100),TUM(100,4),DEN(100,4),Q,QBAR,ZCUR,ZCUR2,QNEC	0062
000010	COMMON/RAW/W(100),Y(100),M,PHASED(100),ALAMDA,FR,WTM(100),	0063
	C WTP(100),KTYP	0064
000010	COMMON/PAW1/ICALL,KCALL,LIM	
000010	QNEC=CMPLX(1.00,0.00)	0066
000012	PI=-3.14159265358979	0067
000014	K=N/4	0068
000015	IF(ICALL.NE.0)GOTO101	0069
000015	DO 102 I=1,M	0070
000020	102 Z(I)=(EXP(CMPLX(0.00,W(I)*PI))	0071
000040	101 A1=0.00	0072
000041	A2=0.00	0073
000042	DO 40 I=1,M	0074
000043	ZCUR=Z(I)	0075
000044	ZCUR2=ZCUR*ZCUR	0076
000053	Q=CMPLX(1.00,0.00)	0077
000056	DO 34 J=1,K	0078
000057	J4=(J-1)*4	0079
000061	TUM(I,J)=1.00+X(J4+1)*ZCUR+X(J4+2)*ZCUR2	0080
000063	DEN(I,J)=1.00+X(J4+3)*ZCUR+X(J4+4)*ZCUR2	0081
000120	33 Q=Q*TUM(I,J)/DEN(I,J)	0082
000147	QPAR=CONJG(Q)	0083
000152	YHT(I)=Q*QPAR	0084
000150	A2=A2+YHT(I)*WTM(I)	0085
000163	YHT(I)=SQRT(YHT(I))	0086
000167	40 A1=A1+YHT(I)*Y(I)*WTM(I)	0087
000201	IF(KTYP.NE.0) GOTO 666	0088
000202	A=A1/A2	0089
000204	GOTO 667	0090
000205	666 A=1.	0091
000207	667 CONTINUE	0092
000207	DO 612 I=1,M	0093
000211	ZZCUR=Z(I)	0094
000214	ZZCUR2=ZZCUR*ZZCUR	0095
000221	ZC=CMPLX(1.,0.)	0096
000224	PHAS=0.	0097
000225	DO 611 J=1,K	0098
000226	J47=(J-1)*4	0099
000230	ZQBAR=QNEC+X(J47+1)*ZZCUR+X(J47+2)*ZZCUR2	0100
000247	ZA1=ZQBAR	0101
000251	ZA2=CMPLX(0.,-1.)*ZQBAR	0102
000261	PHAS=PHAS+ATAN2(ZA2,ZA1)	0103
000265	ZQ=ZQ*ZQBAR	0104
000273	ZQBAR=QNEC+X(J47+3)*ZZCUR+X(J47+4)*ZZCUR2	0105
000315	ZA1=ZQBAR	0106
000317	ZA2=CMPLX(0.,-1.)*ZQBAR	0107
000327	PHAS=PHAS-ATAN2(ZA2,ZA1)	0108
000333	ZC=ZC/ZQBAR	0109
000342	611 CONTINUE	0110
000350	PHA(I)=-PHAS/PI	0111
000352	612 PHE(I)=PHA(I)-PHASED(I)	0112
000360	DO 57 J=1,16	0113
000361	57 G(J)=0.00	0114
000364	V1=0.	0115
000365	V2=0.	0116
000365	DO 42 I=1,M	0117
000367	ZCUR=Z(I)	0118
000372	ZCUR2=ZCUR*ZCUR	0119
000377	YHT(I)=A*YHT(I)	0120
000402	F(I)=YHT(I)-Y(I)	0121
000405	V1=V1+F(I)*E(I)*WTM(I)	0122
000410	V2=V2+F(I)*PHE(I)*WTM(I)	0123
000414	H=F(I)*WTM(I)	0124
000416	T=PHE(I)*WTP(I)*2.*ALAMDA	0125

APPENDIX - Continued

```

000422      DO 422 J=1,K                                0126
000423      J4=(J-1)*4                                    0127
000425      I=2.*4*YHT(I)/TUM(I,J)                      0128
000441      G(J4+1)=G(J4+1)+J*ZCUR                     0129
000451      G(J4+2)=G(J4+2)+0*7CUR2                    0130
000451      Q=-2.*4*YHT(I)/DEN(I,J)                    0131
000475      G(J4+3)=G(J4+3)+J*ZCUR                     0132
000505      G(J4+4)=G(J4+4)+0*7CUR2                     0133
000514      422 CONTINUE                                  0134
000514      DO 42 J=1,K                                  0135
000520      J4=(J-1)*4                                    0136
000522      G(J4+1)=G(J4+1)+T*AIMAG(ZCUR/TUM(I,J))      0137
000527      G(J4+2)=G(J4+2)+T*AIMAG(7CUR2/TUM(I,J))     0138
000554      G(J4+3)=G(J4+3)+T*AIMAG(-7CUR/DEN(I,J))    0139
000571      G(J4+4)=G(J4+4)+T*AIMAG(-7CUR2/DEN(I,J))   0140
000594      42 CONTINUE                                  0141
000613      F=V1+ALAMDA*V2                                0142
000616      ICALL=ICALL+1                                  0143
000620      KCALL=KCALL+1
000621      IF(KCALL.GT.700)RETURN
000624      IF((ICALL/10)*10.EQ.ICALL-1)WRITE(6,2F)ICALL,F,(G(J),J=1,N) 0144
000657      2F FORMAT(* CALL NO.%,I4,* F=*,F15.8/( 25X, 4F15.8)) 0145
000657      IF(ICALL.GT.2000) GO TO 450
000663      IF(ICALL.GT.0)RETURN                                0147
000665      GO TO 449
000666      450 IF(ICALL.GT.LIM+1) GO TO 449
000673      RETURN
C.....PRINT OUT
000673      449 WRITE(6,50)F                                0148
000701      50 FORMAT(* FINAL FUNCTION VALUE =*,F15.8)    0149
000701      WRITE(6,51)A                                     0150
000707      51 FORMAT(* A=*,F15.8)                        0151
000707      WRITE(6,52)(X(J),J=1,N)                       0152
000731      52 FORMAT(* FINAL X =*/(* *,4F15.8))         0153
000731      WRITE(6,54)(G(J),J=1,N)                       0154
000753      54 FORMAT(* FINAL GRADIENT =*/(* *,4F15.8))  0155
000753      DO 55 I=1,M                                    0157
000760      55 WRITE(6,55)I,W(I),Y(I),YHT(I),E(I),PHASE(I),PHA(I),PHE(I) 0158
000760      56 FORMAT(* I=*,I2,* W=*,F8.5,* Y=*,F8.5,* YHT=*,F8.5,* E=*,F8.5, 0159
000760      C * PHASE=*,F8.5,* PHA=*,F8.5,* PHE=*,F8.5) 0160
001011      WRITE(6,59)K                                    0161
001016      59 FORMAT(* FINAL TABLE FOR A*,I2,* STAGE FILTER*) 0162
001016      YMIN33=0.                                       0163
001017      YMAX33=0.                                       0164
001020      YMAX11=0.                                       0165
001021      S=FR/200.                                       0166
001023      DO 60 I=1,201                                   0167
001027      FREQ=S*FLOAT(I-1)                             0168
001032      ZCUR=CEXP(CMPLX(0.00,FREQ*PI))               0169
001040      ZCUR2=ZCUR*ZCUR                                0170
001045      Q=CMPLX(1.00,C.00)                             0171
001051      PHASE=0.00                                      0172
001051      DO 61 J=1,K                                    0173
001055      J4=(J-1)*4                                    0174
001057      QBAR=ONE0+X(J4+1)*ZCUR+X(J4+2)*ZCUR2         0175
001076      A1=QBAR                                          0176
001100      A2=CMPLX(0.00,-1.00)*QBAR                     0177
001110      PHASE=PHASE+ ATAN2(A2,A1)                     0178
001114      Q=Q*QBAR                                         0179
001122      QBAR=ONE0+X(J4+2)*ZCUR+X(J4+4)*ZCUR2        0180
001144      A1=QBAR                                          0181
001146      A2=CMPLX(0.00,-1.00)*QBAR                     0182
001156      PHASE=PHASE- ATAN2(A2,A1)                     0183
001162      61 Q=Q/QBAR                                       0184
001177      A1=Q* CONJG(Q)                                   0185
001207      A1=A* SQRT(A1)                                   0186
001212      PHASE=-PHASE/PI                                0187
001214      CMFGA(I)=FREQ                                    0188
001216      AMAG(I)=A1                                       0189
001217      PHASEX(I)=PHASE                                0190

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APPENDIX - Continued

001221	IF (PHASE-YMAX33) 301,301,401	0191
001226	401 YMAX23=PHASE	0192
001230	301 CONTINUE	0193
001230	IF (PHASE-YMIN33) 402,302,302	0194
001233	402 YMIN33=PHASE	0195
001235	302 CONTINUE	0196
001235	IF (A1-YMAX11) 300,300,400	0197
001240	400 YMAX11=A1	0198
001242	300 CONTINUE	0199
001247	60 WRITE (5,62) FREQ, PHASE, A1	0200
001261	62 FORMAT (* W=*, F15.8, * PHASE/PI=*, F15.8, * YHT=*, F15.8)	0201
	C SCALE COMPUTATIONS	0202
001261	IYMAX11=YMAX11+.9999	0203
001264	YMAX1=FLOAT(IYMAX11)	0204
001266	IF (HMAX.EQ.0.) GO TO 333	0205
001267	YMAX1=HMAX	0206
001270	333 CONTINUE	0207
001270	IYMAX33=YMAX33+.999	0208
001273	YMAX3=FLOAT(IYMAX33)	0209
001275	IYMIN33=YMIN33+.9999	0210
001300	YMIN3=FLOAT(IYMIN33)	0211
001301	AXM=10F F/(FS/2)	0212
001303	AYM=10H MAGNITUDES	0213
001304	AFR=200.*FR	0214
001306	NFR=200	0215
	C MAGNITUDE (COMPUTED) - FREQUENCY PLOT	0216
001307	CALL INFOPLT(C,NFR,CMEGA,1,AMAG,1,C.,FR,C.,YMAX1,C.5,-10,AXM,-10,	0217
	C AYM,0)	0218
	C MAGNITUDE (DESIRED) - FREQUENCY PLOT	0219
001327	CALL INFOPLT(1,M.W,1,Y,1, C.,FR,C.,YMAX1,C.5,-10,AXM,-10,AYM,11)	0220
001347	AYM=10H PHASE/PI	0221
	C PHASE-FREQUENCY PLOT	0222
001351	CALL INFOPLT(C,NFR,CMEGA,1,PHASEX,1,C.,FR,YMIN3,YMAX3,C.5,-10,	0223
	C AXM,-10,AYM,0)	0224
	C PHASE (DESIRED) - FREQUENCY PLOT	0225
001370	CALL INFOPLT(1,M.W,1,Y,1, C.,FR,C.,YMIN3,YMAX3,C.5,-10,	0226
	C AXM,-10,AYM,11)	0227
001410	RETURN	0228
001411	END	0229
	SUBROUTINE FLPLW (FUNCT,N,X,F,G,EST,EPS,LIMIT,IER,H)	0230
000015	DIMENSION H(1),X(1),G(1)	0231
000015	COMMON/RAW1/ICALL,KCALL,LIM	
000015	CALL FUNCT(N,X,F,G,HMAX)	0233
000027	IF (KCALL.GT.300) GO TO 723	
000032	IF (ICALL.GT.LIM) GO TO 723	
000035	GO TO 517	
000035	723 IER=3	
000037	RETURN	
000037	517 IER=0	
000040	KCOUNT=0	0235
000041	N2=N+N	0236
000042	N3=N2+N	0237
000043	N4=N3+1	0238
000045	K=N3	0239
000047	DO 4 J=1,N	0240
000050	H(K)=1.00	0241
000053	NJ=N-1	0242
000054	IF (NJ) 5,5,2	0243
000055	DO 3 L=1,NJ	0244
000060	KI=K+L	0245

APPENDIX - Continued

000061	3	H(KL)=0.00	0246
000066	4	K=KL+1	0247
000072	5	KCOUNT=KCUNT+1	0248
000074		WRITE(5,501)KCUNT	0249
000101	501	FORMAT(*,KOUNT=*,15)	0250
000101		QLDF=F	0251
000106		DO 5 J=1,N	0252
000107		K=N+J	0253
000110		H(K)=G(J)	0254
000114		K=K+N	0255
000115		H(K)=X(J)	0256
000121		K=J+N3	0257
000122		T=C.00	0258
000123		DO 4 I=1,N	0259
000124		T=T-G(I)*H(K)	0260
000131		IF(L-J) 5,7,7	0261
000134	6	K=K+N-L	0262
000137		GO TO 4	0263
000137	7	K=K+1	0264
000141	8	CONTINUE	0265
000144	9	H(J)=T	0266
000150		DY=C.00	0267
000150		HNRM=0.00	0268
000151		GNRM=0.00	0269
000153		DO 10 J=1,N	0270
000154		HNRM=HNRM+ABS(H(J))	0271
000160		GNRM=GNRM+ABS(G(J))	0272
000163	10	DY=DY+H(J)*G(J)	0273
000173		IF(DY) 11,51,51	0274
000174	11	IF(HNRM/GNRM-EPS) 51,51,12	0275
000200	12	FY=F	0276
000201		ALFA=2.00*(EST-F)/DY	0277
000204		AMBDA=1.00	0278
000205		IF(ALFA) 15,15,13	0279
000207	13	IF(ALFA-AMBDA) 14,15,15	0280
000212	14	AMBDA=ALFA	0281
000214	15	ALFA=0.00	0282
000215	16	FX=FY	0283
000216		DX=DY	0284
000220		DO 17 I=1,N	0285
000222	17	X(I)=X(I)+AMBDA*H(I)	0286
000231		CALL FUNCT(N,X,F,G,HMAX)	0289
000243		IF(KCALL.GT.300) GO TO 724	
000246		IF(ICALL.GT.LIM) GO TO 724	
000251		GO TO 918	
000251	724	IER=3	
000253		RETURN	
000253	918	FY=F	
000254		DY=C.00	0292
000255		DO 18 I=1,N	0293
000257	18	DY=DY+G(I)*H(I)	0294
000266		IF(DY) 19,26,22	0297
000267	19	IF(FY-FX) 20,22,22	0298
000272	20	AMBDA=AMBDA+ALFA	0299
000274		ALFA=AMBDA	0300
000275		ERROR=1.E10	0301
000276		IF(HNRM*AMBDA-ERROR) 16,16,21	0302
000302	21	IER=2	0303
000304		RETURN	0304
000304	22	T=0.00	0305
000305	23	IF(AMBDA) 24,36,24	0306
000306	24	Z=3.00*(FX-FY)/AMBDA+DX+DY	0307
000314		ALFA=AMAX1(ABS(Z),ABS(DX),ABS(DY))	0308
000326		CALFA=Z/ALFA	0309

APPENDIX - Continued

000327		DALFA=DALFA*DALFA-DX/ALFA*DY/ALFA	0310
000333		IF(DALFA) 51,25,25	0311
000335	25	W=ALFA* SQRT(CALFA)	0312
000340		ALFA= (DY+W-Z)*AMBDA/(DY+2.CO*W-DX)	0313
000351		DO 26 I=1,N	0314
000356	26	X(I)=X(I)+(T-ALFA)*H(I)	0315
000366		CALL FUNCT(N,X,F,G,H*AX)	0320
000400		IF(KCALL.GT.300) GO TO 725	
000403		IF(ICALL.GT.LIM) GO TO 725	
000406		GO TO 919	
000406	725	IER=3	
000410		RETURN	
000410	919	IF(F-FX) 27,27,28	
000413	27	IF(F-FY) 36,36,28	0323
000416	28	DALFA=0.00	0324
000417		DO 29 I=1,N	0325
000421	29	DALFA=DALFA+G(I)*H(I)	0326
000430		IF(DALFA) 30,33,33	0329
000431	30	IF(F-FX) 32,31,33	0330
000434	31	IF(DX-DALFA) 32,36,32	0331
000436	32	FX=F	0332
000437		DX=DALFA	0333
000440		T=ALFA	0334
000442		AMBDA=ALFA	0335
000443		GO TO 23	0336
000443	33	IF(FY-F) 35,34,35	0337
000445	34	IF(DY-DALFA) 35,36,35	0338
000447	35	FY=F	0339
000450		DY=DALFA	0340
000451		AMBDA=AMBDA-ALFA	0341
000454		GO TO 22	0342
000454	36	DO 37 J=1,N	0343
000456		K=N+J	0344
000457		H(K)=G(J)-H(K)	0345
000463		K=K+N	0346
000464	37	H(K)=X(I)-H(K)	0347
000472		IF(OLDF-F+EPS) 51,38,38	0348
000476	38	IER=0	0349
000477		IF(KOUNT-N) 42,39,39	0350
000501	39	T=0.00	0351
000502		Z=0.00	0352
000503		DO 40 J=1,N	0353
000504		K=N+J	0354
000505		W=H(K)	0355
000510		K=K+N	0356
000511		T=T+ APS(H(K))	0357
000514	40	Z=Z+W*H(K)	0358
000523		IF(HNRN-EPS) 41,41,42	0359
000526	41	IF(T-EPS) 56,56,42	0360
000531	42	IF(KOUNT-LIMIT) 43,50,50	0361
000534	43	ALFA=C.C	0362
000535		DO 47 J=1,N	0363
000537		K=J+N3	0364
000540		W=C.00	0365
000542		DO 44 L=1,N	0366
000543		KL=N+L	0367
000544		W=W+H(KL)*H(K)	0368
000552		IF(L-J) 44,45,45	0369
000554	44	K=K+N-L	0370
000557		GO TO 45	0371
000557	45	K=K+1	0372
000561	46	CONTINUE	0373
000564		K=N+J	0374
000565		ALFA=ALFA+W*H(K)	0375

APPENDIX -- Continued

000572	47	H(J)=W	0376
000576		IF(Z*ALFA) 48,1,48	0377
000600	48	K=N2	0378
000602		DO 49 L=1,N	0379
000603		KL=N2+L	0380
000605		DO 49 J=L,N	0381
000606		NJ=N2+J	0382
000607		H(K)=H(K)+H(KL)*H(NJ)/Z-H(L)*H(J)/ALFA	0383
000625	49	K=K+1	0384
000633		GO TO 5	0385
000634	50	IFR=1	0386
000636		RETURN	0387
000636	51	DO 52 J=1,N	0388
000640		K=N2+J	0389
000641	52	X(J)=H(K)	0390
000647		CALL FUNCT(N,X,F,G,HMAX)	0393
000661		IF(KCALL.GT.300) GO TO 726	
000664		IF(ICALL.GT.LIM) GO TO 726	
000667		GO TO 920	
000667	726	IFR=3	
000671		RETURN	
000671	920	IF(GNRM-EPS) 55,55,57	
000674	53	IF(IFR) 56,54,54	0396
000676	54	IFR=-1	0397
000700		GO TO 1	0398
000700	55	IFR=0	0399
000701	56	RETURN	0400
000702		END	0401

```

SUBROUTINE INSIDE(N,X,KFLAG)
  DIMENSION X(16)
  J=-1
  KFLAG=0
10 J=J+2
  IF(J.GT.N)RETURN
  B=-.500*X(J)
  C=X(J+1)
  DISC=B*B-C
  IF(DISC.LE.0.00)GOTO20
C.....REAL ROOTS
  DISC= SQRT(DISC)
  R1=B+DISC
  R2=B-DISC
  CR1= ABS(R1)
  CR2= ABS(R2)
  IF(DR1.LE.1.00.AND.DR2.LE.1.00)GOTO10
  KFLAG=1
  IF(DR1.GT.1.00)R1=1.00/R1
  IF(DR2.GT.1.00)R2=1.00/R2
  X(J)=-1.00*(R1+R2)
  X(J+1)=R1*R2
  GOTO10
C.....COMPLEX ROOTS
20 IF(C.LE.1.00)GOTO10
  KFLAG=1
  C=1.00/C
  X(J+1)=C
  X(J)=X(J)*C
  GOTO10
END

```

APPENDIX – Concluded

000005	SUBROUTINE ROOTS(N,X)	0433
000005	DIMENSION X(16)	0434
000005	WRITE(6,40)	0435
000010	40 FORMAT(* ROOTS*/6X,*REAL*,11X,*IMAG*,11X,*REAL*,11X,*IMAG*)	0436
000010	J=-1	0437
000011	10 J=J+2	0438
000013	IF(J.GT.N) RETURN	0439
000017	B=-.500*X(J)	0440
000021	C=X(J+1)	0441
000023	DISC=B*B-C	0442
000025	IF(DISC.LE.0.00) GOTO 20	0443
	C.....REAL RCOTS	0444
000027	DISC= SQRT(DISC)	0445
000030	R1=B+DISC	0446
000032	R2=B-DISC	0447
000035	WRITE(6,30) R1,R2	0448
000044	30 FORMAT(* *,F15.8,15X,F15.8)	0449
000044	GOTO 10	0450
	C.....COMPLEX ROOTS	0451
000046	20 DISC= SQRT(-1.00*DISC)	0452
000052	DISCM=-1.00*DISC	0453
000055	WRITE(6,50) B,DISC,B,DISCM	0454
000070	50 FORMAT(* *,4F15.8)	0455
000070	GOTO 10	0456
000072	END	0457

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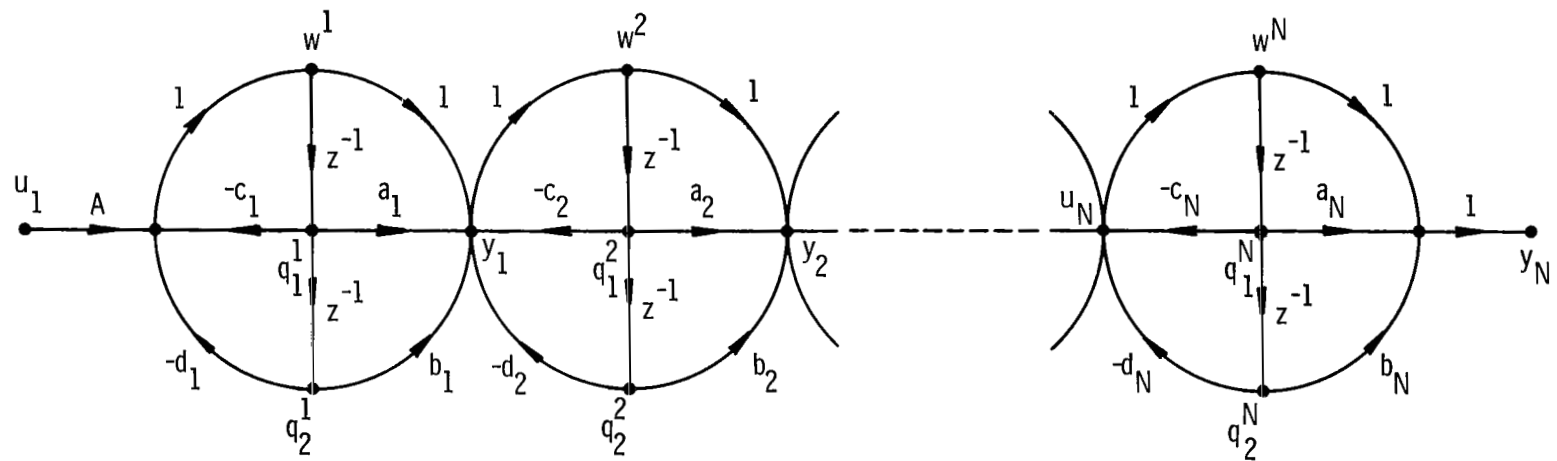


Figure 1.- Signal flow graph of cascaded digital filter.

$$\theta_k = 1 - 2\Omega_k \quad (0 \leq \Omega_k \leq 1)$$

$$M_k = \text{Unspecified} \quad (0 \leq \Omega_k \leq 1)$$

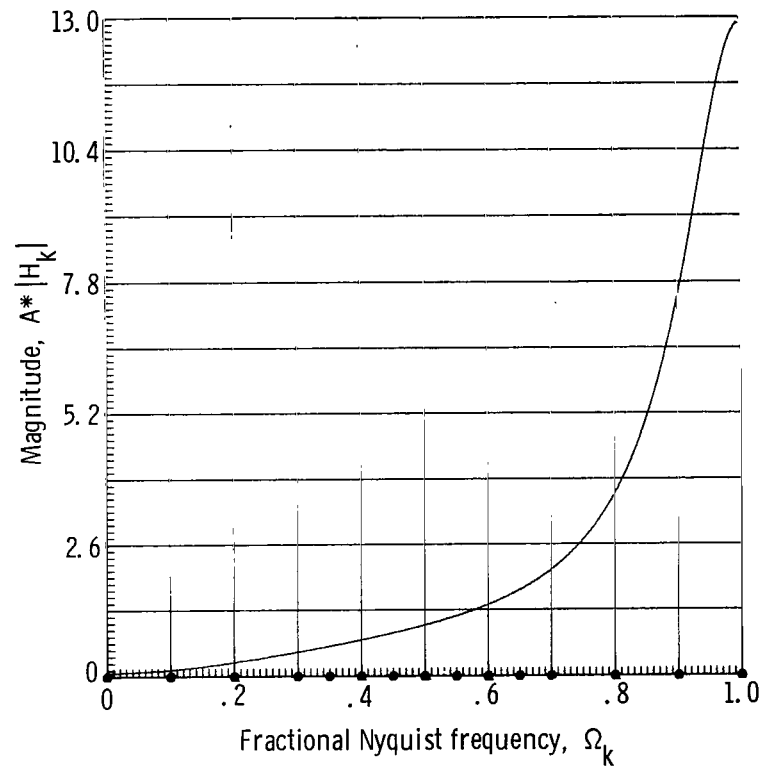
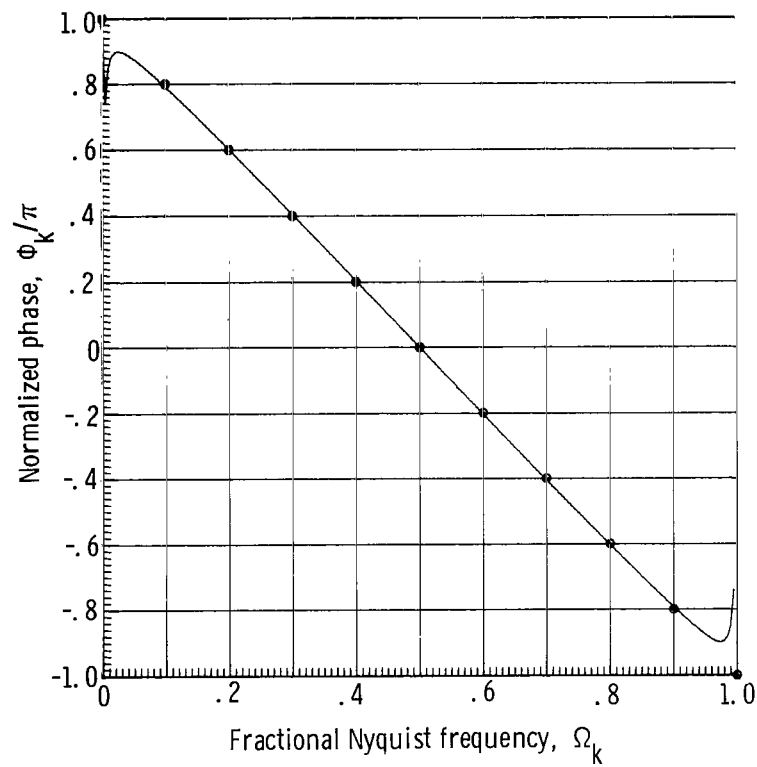
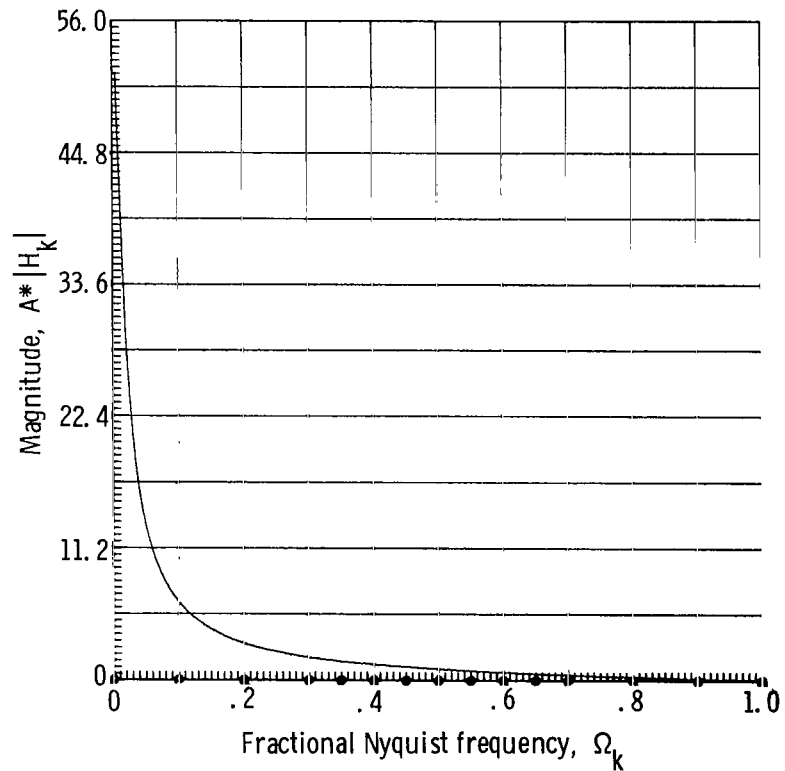
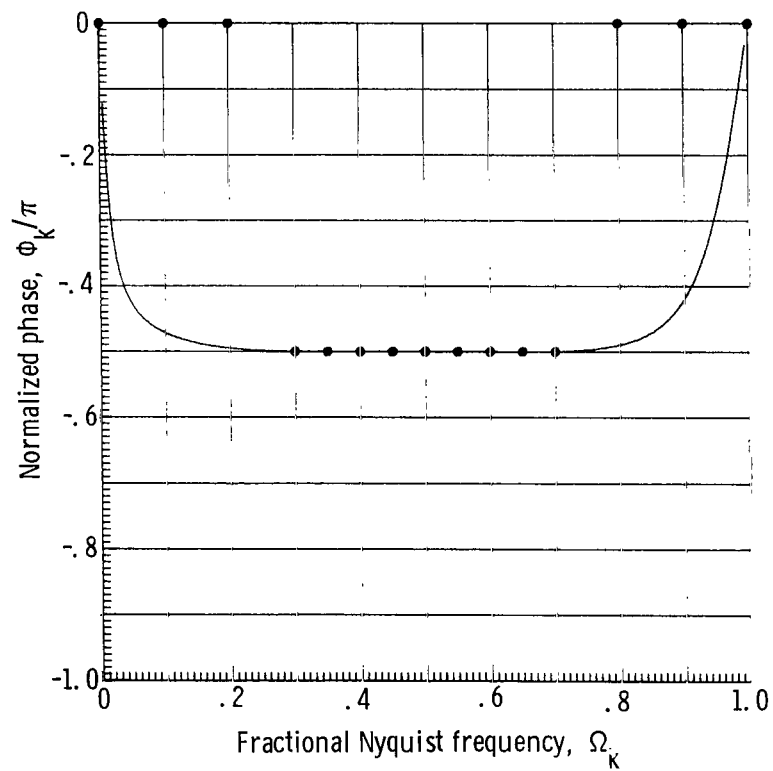


Figure 2.- Two-stage linear-phase filter.

$$\theta_k = \begin{cases} -\pi/2 & (0.3 \leq \Omega_k \leq 0.7) \\ \text{Unspecified} & (\text{elsewhere}) \end{cases}$$

$$M_k = \text{Unspecified} \quad (0 \leq \Omega_k \leq 1)$$

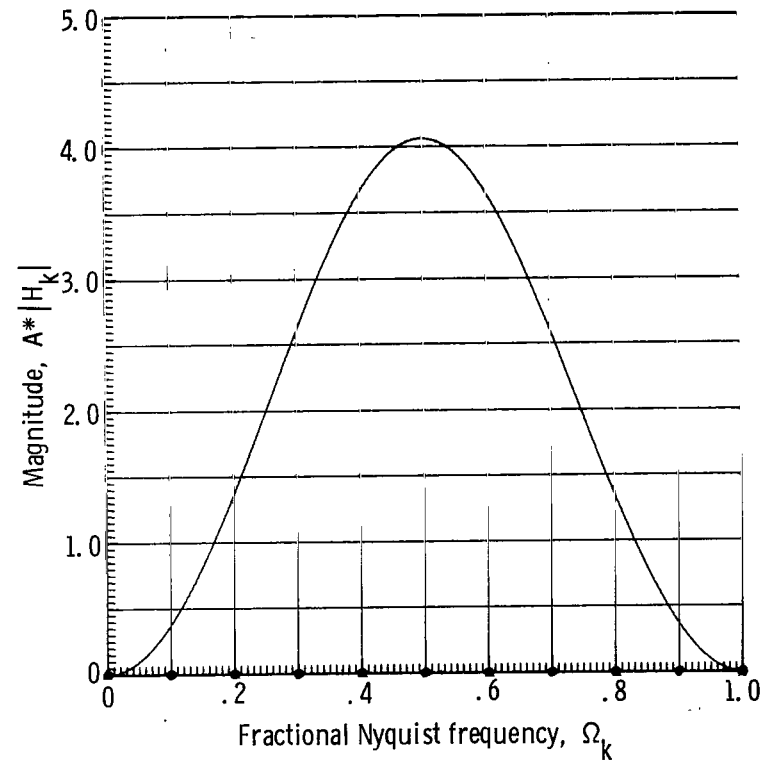
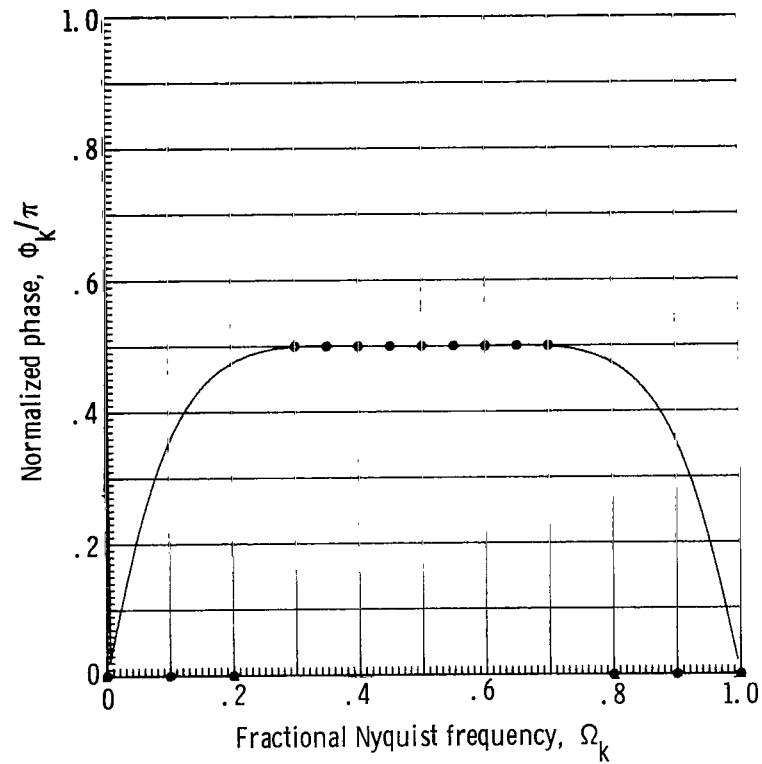


(a) Lag filter.

Figure 3.- Two-stage constant-phase filters.

$$\theta_k = \begin{cases} \pi/2 & (0.3 \leq \Omega_k \leq 0.7) \\ \text{Unspecified} & (\text{elsewhere}) \end{cases}$$

$$M_k = \text{Unspecified} \quad (0 \leq \Omega_k \leq 1)$$

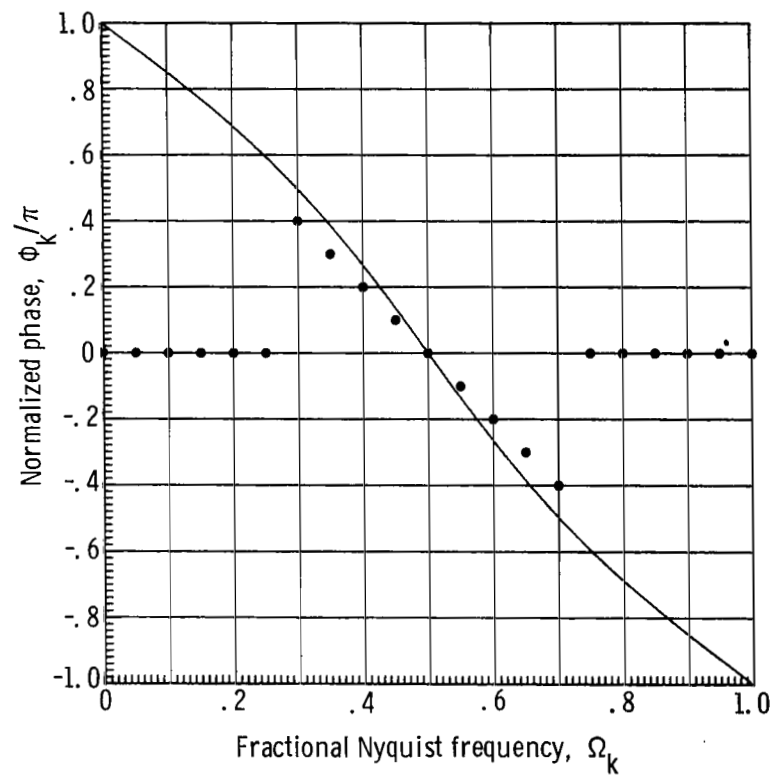
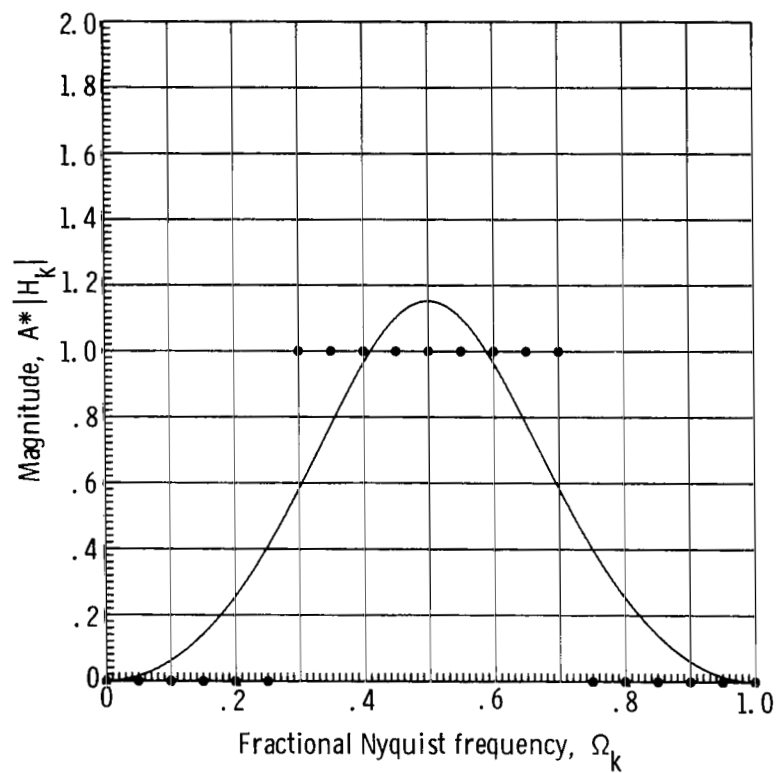


(b) Lead filter.

Figure 3.- Concluded.

$$M_k = \begin{cases} 1 & (0.3 \leq \Omega_k \leq 0.7) \\ 0 & (\text{elsewhere}) \end{cases}$$

$$\theta_k = \text{Unspecified} \quad (0 \leq \Omega_k \leq 1)$$

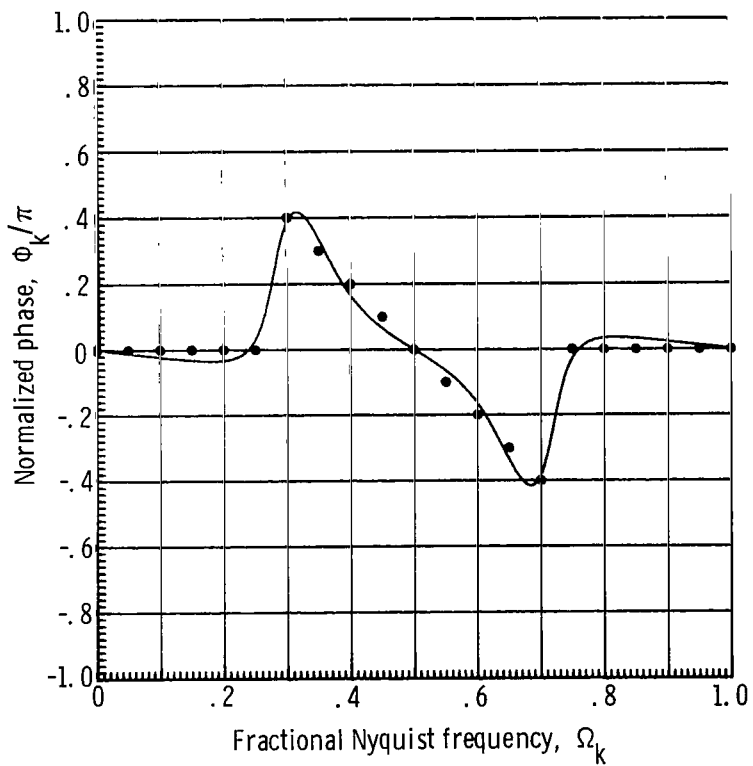
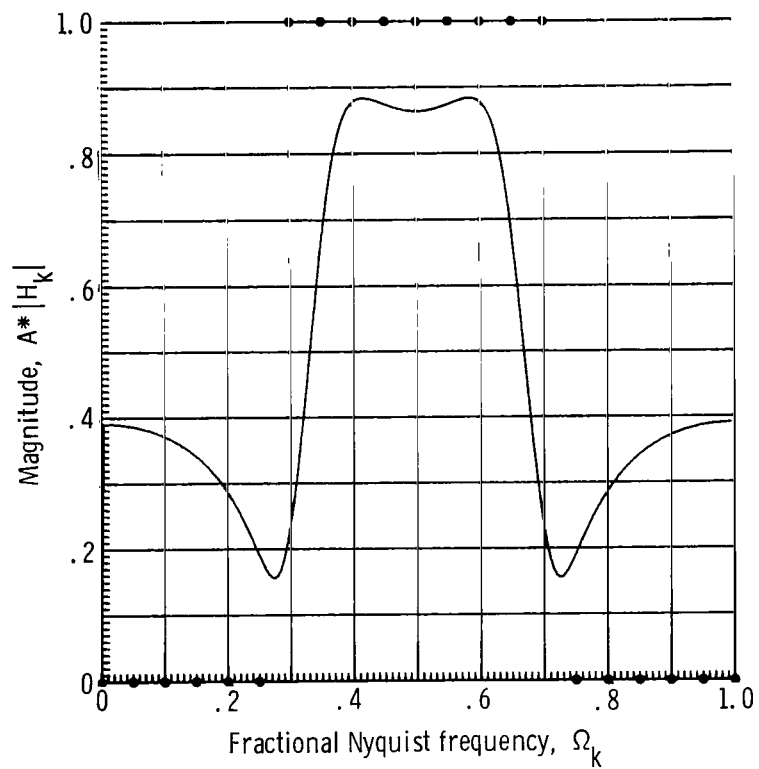


(a) Unspecified phase filter. $\lambda = 0$.

Figure 4.- Two-stage limited-band constant-gain filters.

$$M_k = \begin{cases} 1 & (0.3 \leq \Omega_k \leq 0.7) \\ 0 & (\text{elsewhere}) \end{cases}$$

$$\theta_k = \begin{cases} 1-2\Omega_k & (0.3 \leq \Omega_k \leq 0.7) \\ 0 & (\text{elsewhere}) \end{cases}$$

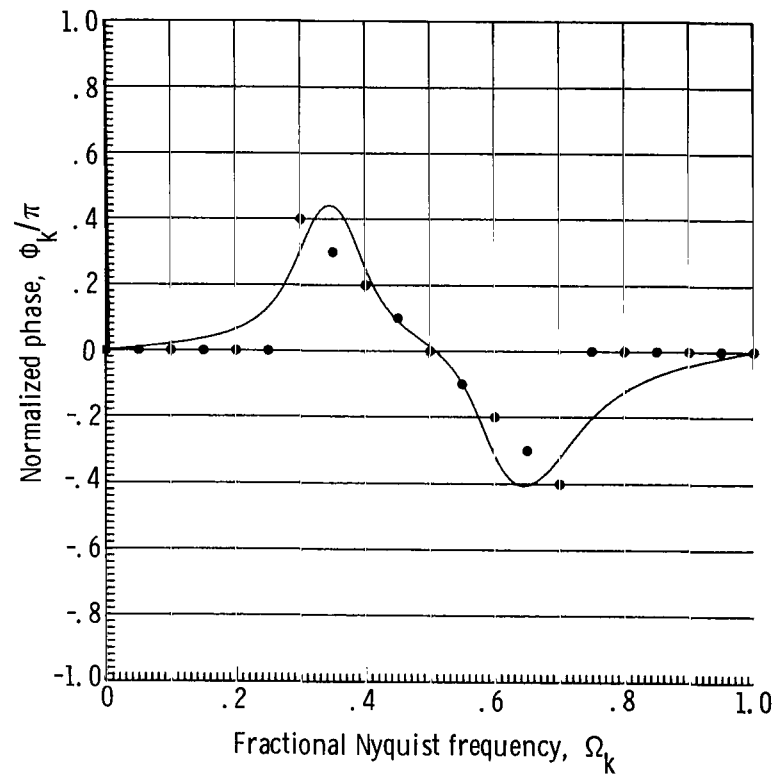
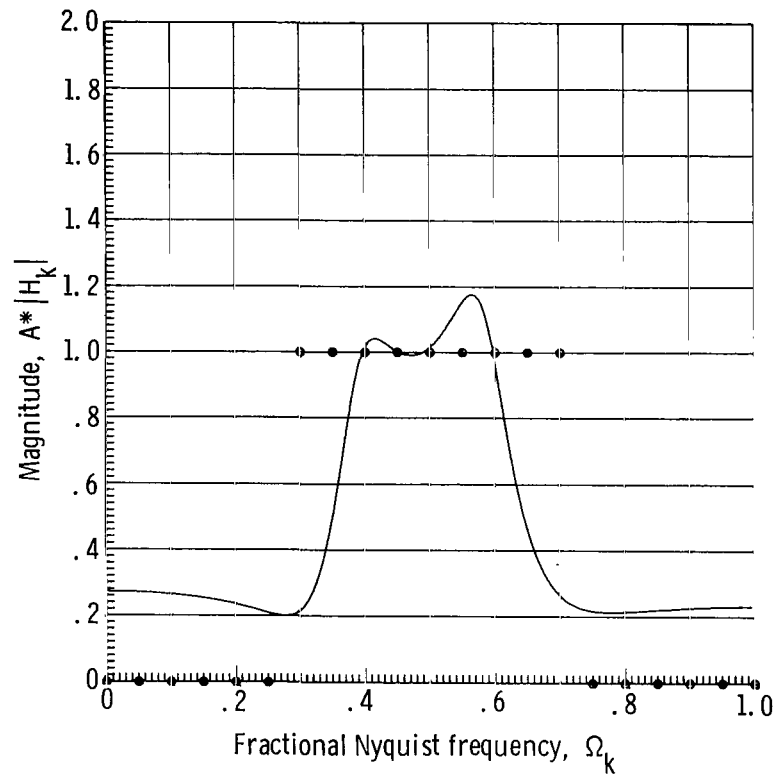


(b) Linear-phase filter. $\lambda = 10$.

Figure 4.- Continued.

$$M_k = \begin{cases} 1 & (0.3 \leq \Omega_k \leq 0.7) \\ 0 & (\text{elsewhere}) \end{cases}$$

$$\theta_k = \begin{cases} 1-2\Omega_k & (0.3 \leq \Omega_k \leq 0.7) \\ 0 & (\text{elsewhere}) \end{cases}$$

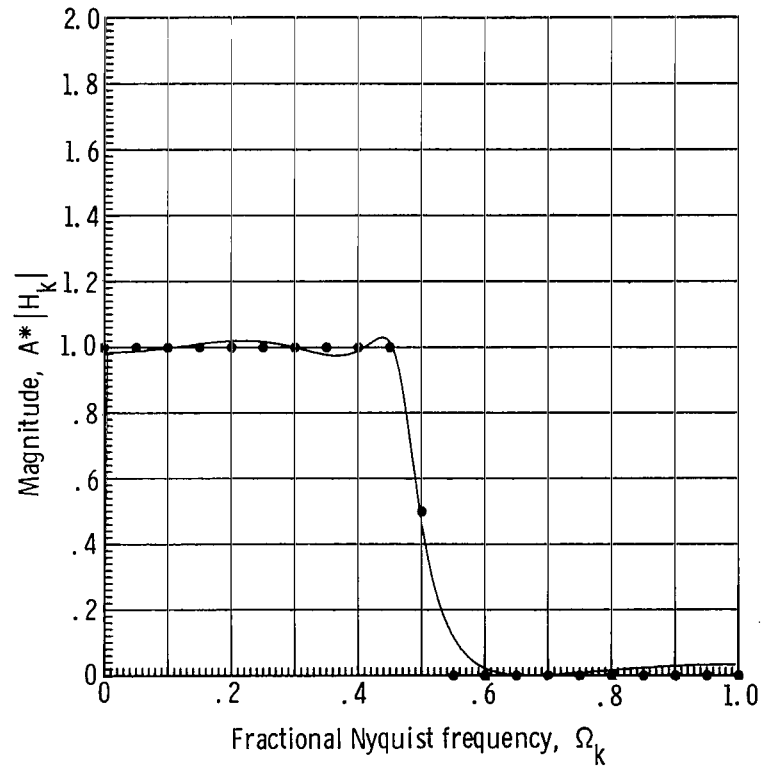
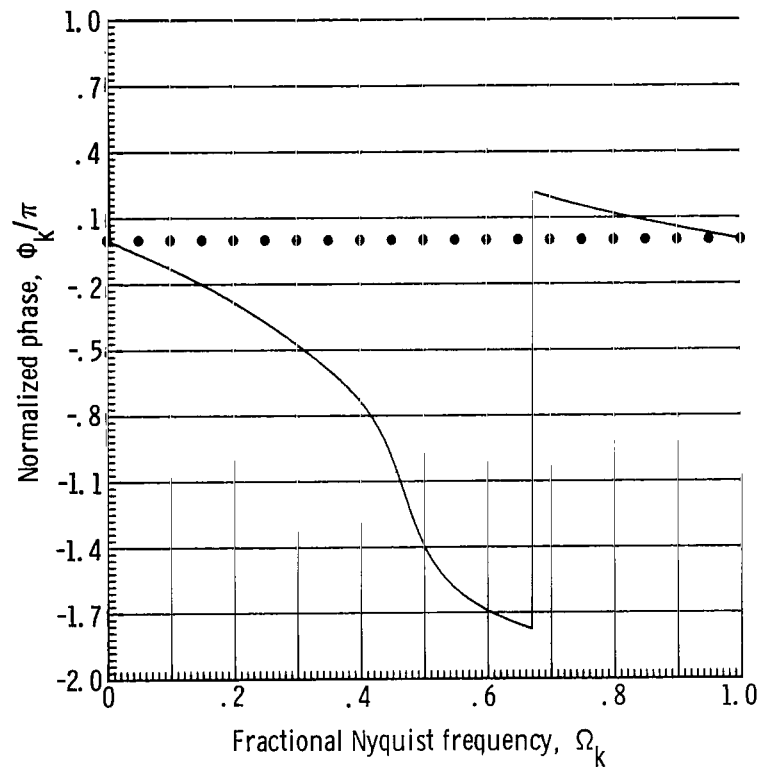


(c) Linear-phase filter. $\lambda = 1000$.

Figure 4.- Concluded.

$$\theta_k = \text{Unspecified} \quad (0 \leq \Omega_k \leq 1)$$

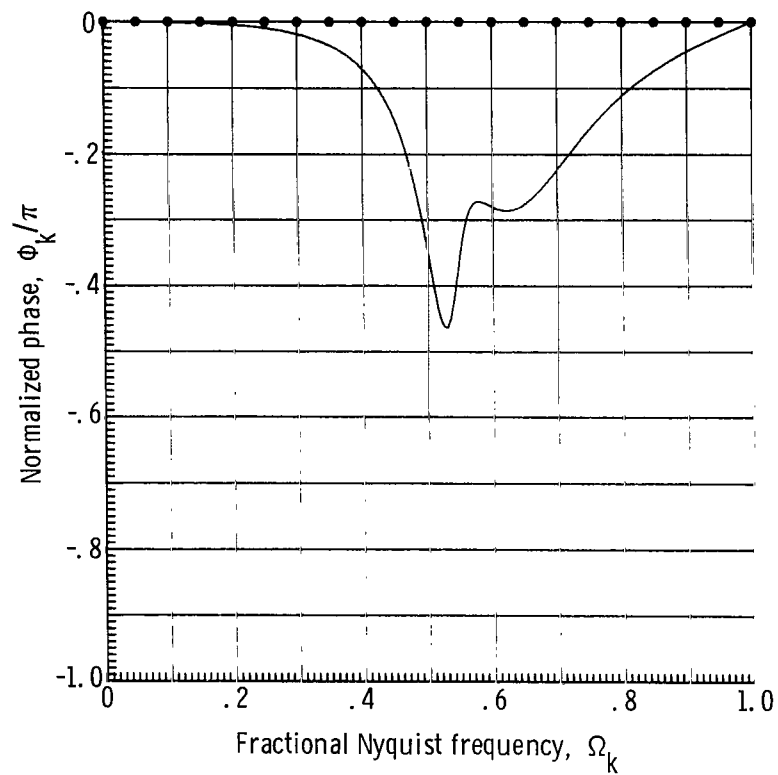
$$M_k = \begin{cases} 1.0 & (0.0 \leq \Omega_k < 0.5) \\ 0.5 & (\Omega_k = 0.5) \\ 0.0 & (0.5 < \Omega_k \leq 1.0) \end{cases}$$



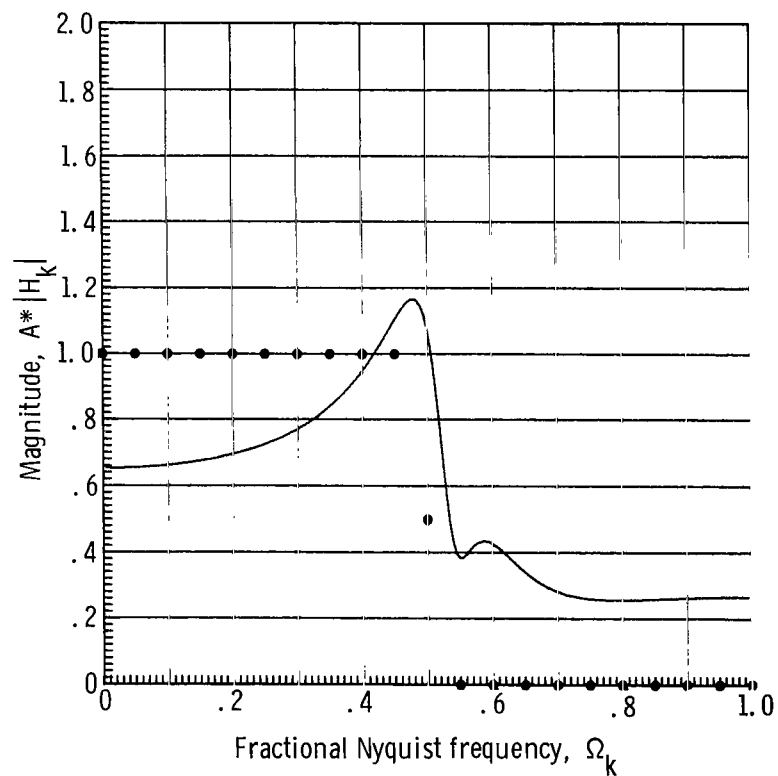
(a) Unspecified-phase filter. $\lambda = 0$.

Figure 5.- Two-stage low-pass filters.

$$\theta_k = \begin{cases} 0 & (0.0 \leq \Omega_k \leq 0.5) \\ \text{Unspecified} & (\text{elsewhere}) \end{cases}$$



$$M_k = \begin{cases} 1.0 & (0.0 \leq \Omega_k < 0.5) \\ 0.5 & (\Omega_k = 0.5) \\ 0.0 & (0.5 < \Omega_k \leq 1.0) \end{cases}$$

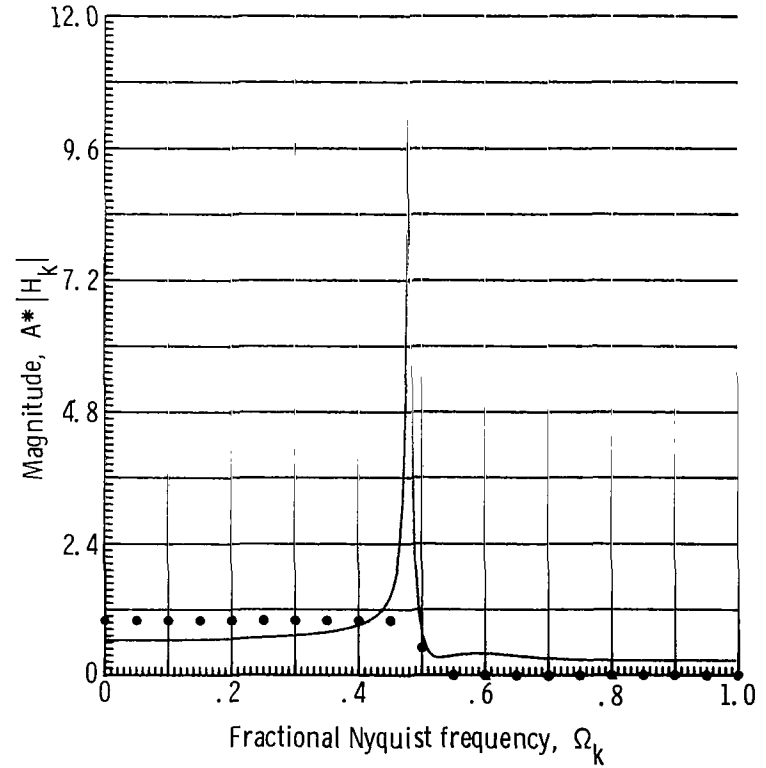
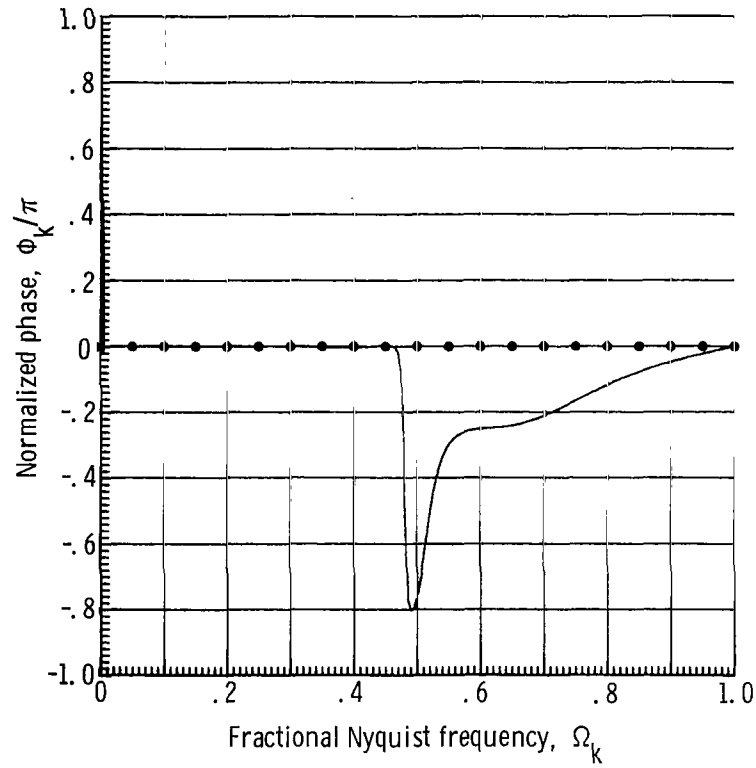


(b) Zero-phase filter. $\lambda = 10$.

Figure 5.- Continued.

$$\theta_k = \begin{cases} 0 & (0.0 \leq \Omega_k \leq 0.5) \\ \text{Unspecified} & (\text{elsewhere}) \end{cases}$$

$$M_k = \begin{cases} 1.0 & (0.0 \leq \Omega_k < 0.5) \\ 0.5 & (\Omega_k = 0.5) \\ 0.0 & (0.5 < \Omega_k \leq 1.0) \end{cases}$$

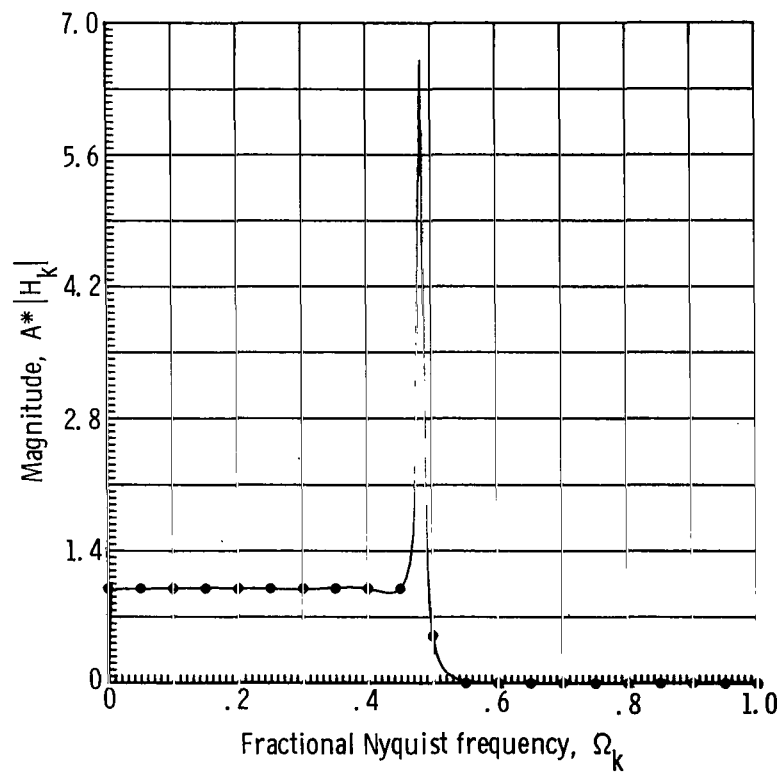
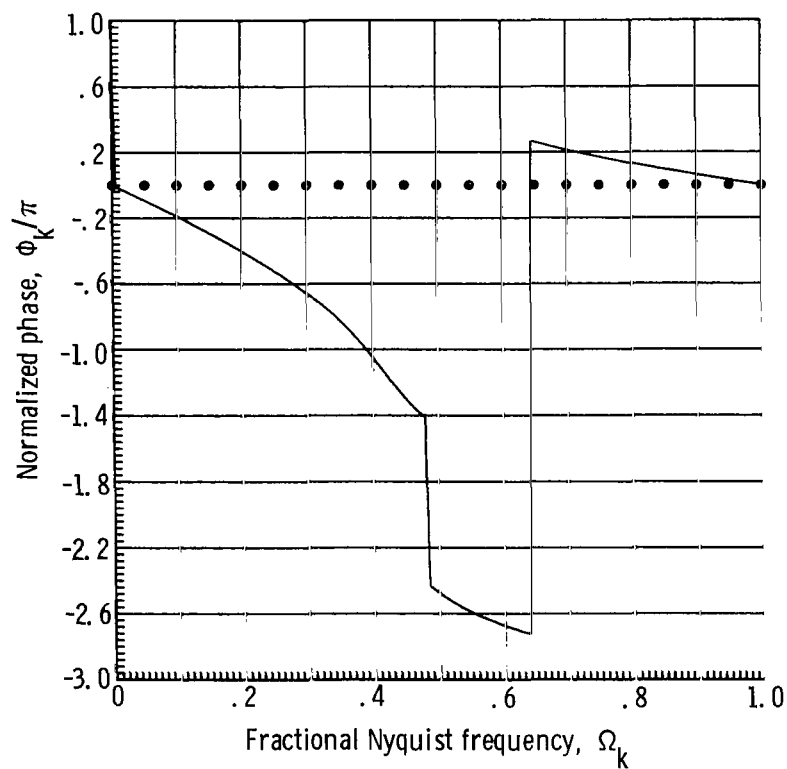


(c) Zero-phase filter. $\lambda = 1000$.

Figure 5.- Concluded.

$\theta_k = \text{Unspecified} \quad (0 \leq \Omega_k \leq 1)$

$$M_k = \begin{cases} 1.0 & (0.0 \leq \Omega_k < 0.5) \\ 0.5 & (\Omega_k = 0.5) \\ 0.0 & (0.5 < \Omega_k \leq 1.0) \end{cases}$$

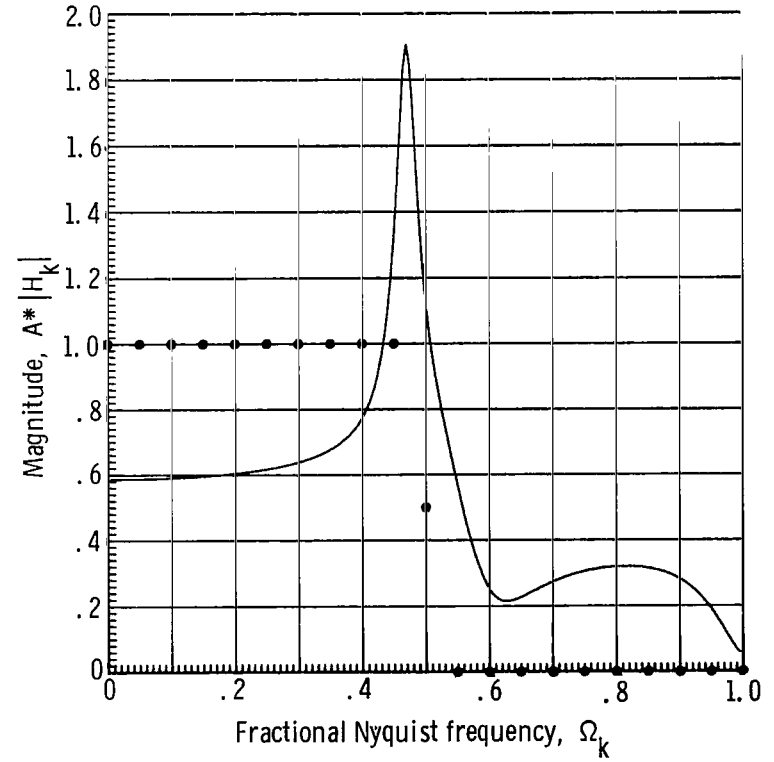
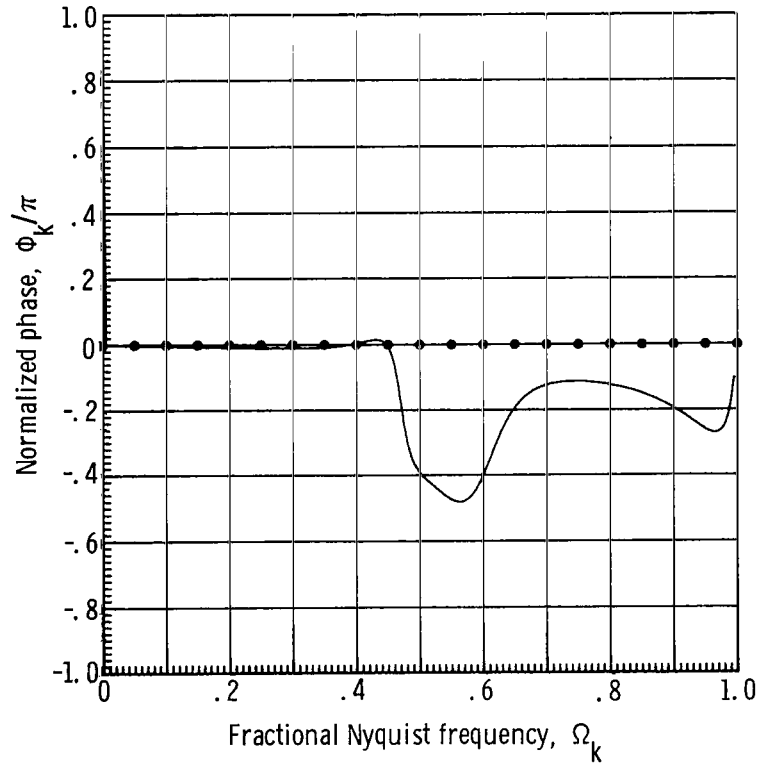


(a) Unspecified-phase filter. $\lambda = 0$.

Figure 6.- Three-stage low-pass filters.

$$\theta_k = \begin{cases} 0 & (0.0 \leq \Omega_k \leq 0.5) \\ \text{Unspecified} & (\text{elsewhere}) \end{cases}$$

$$M_k = \begin{cases} 1.0 & (0.0 \leq \Omega_k < 0.5) \\ 0.5 & (\Omega_k = 0.5) \\ 0.0 & (0.5 < \Omega_k \leq 1.0) \end{cases}$$



(b) Zero-phase filter. $\lambda = 10$.

Figure 6.- Continued.

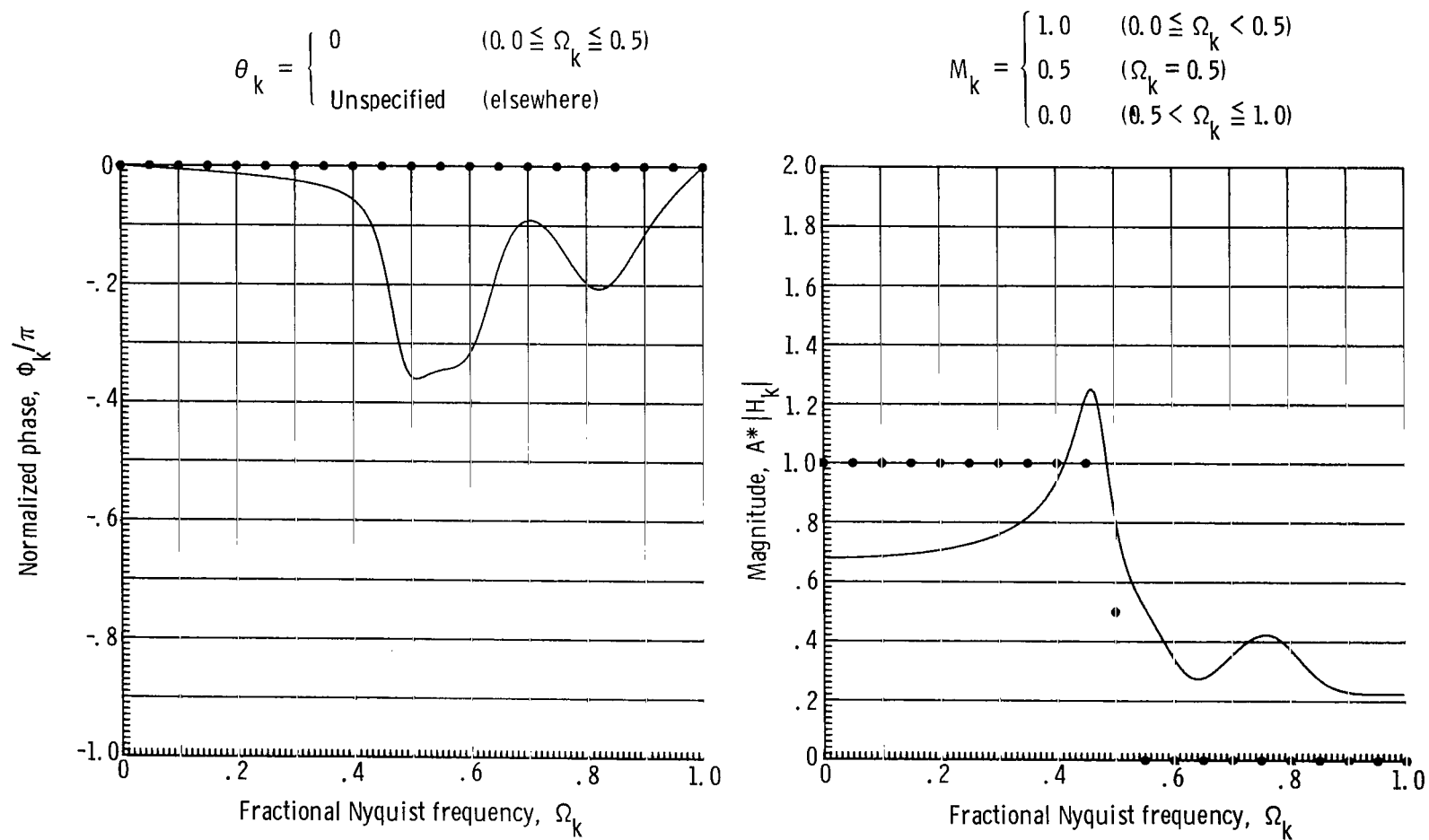
(c) Zero-phase filter. $\lambda = 1000$.

Figure 6.- Concluded.

$$\theta_k = \text{Unspecified} \quad (0 \leq \Omega_k \leq 1)$$

$$M_k = \begin{cases} 1.0 & (0.0 \leq \Omega_k < 0.5) \\ 0.5 & (\Omega_k = 0.5) \\ 0.0 & (0.5 < \Omega_k \leq 1.0) \end{cases}$$

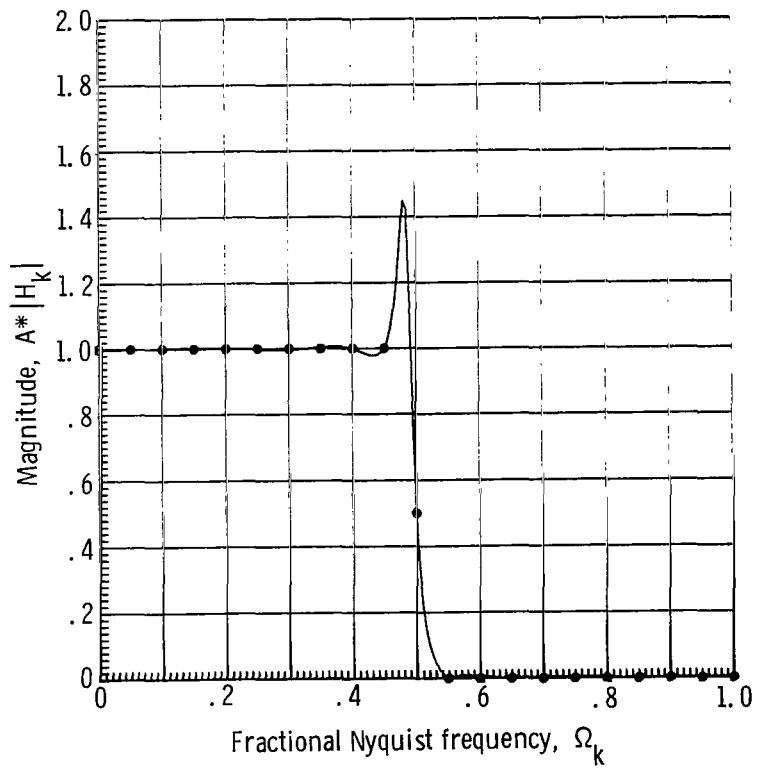
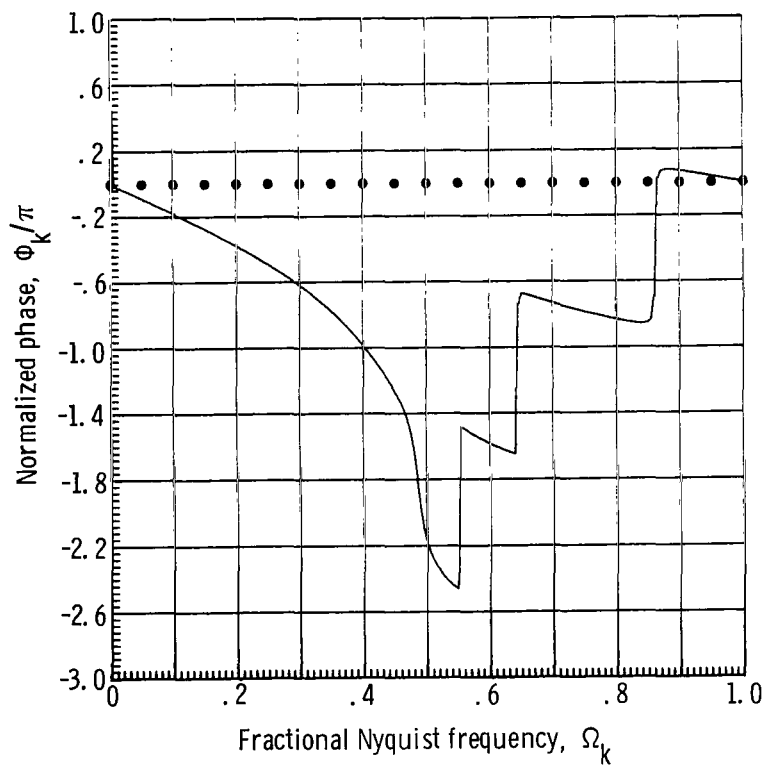


Figure 7.- Three-stage low-pass filter.



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